# Embedded Code Generation from High-level Heterogeneous Components

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# T H E S I S

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# Embedded Code Generation from High-level Heterogeneous Components

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...στους γονείς μου Αθηνά και Δημήτρη

Ithaca

When you set out on your journey to Ithaca,
pray that the road is long,
full of adventure, full of knowledge.
The Lestrygonians and the Cyclops,
the angry Poseidon do not fear
You will never find such as these on your path,
if you do not carry them within your soul,
if your soul does not set them up before you.

And if you find her poor, Ithaca has not deceived you.
Wise as you have become, with so much experience,
you must already have understood what Ithacas mean.

*Constantine P. Cavafy (1911)*

Ιθάκη

Σα βγεις στο πηγαιμό για την Ιθάκη,
να εύχεσαι να είναι μακρύς ο δρόμος,
γεμάτος περιπέτειες, γεμάτος γνώσεις.
Τους Λαιστριγόνας και τους Κύκλωπας,
τον θυμωμένο Ποσειδώνα μη φοβάσαι.
Τέτοια στο δρόμο σου ποτέ δεν θα βρεις,
αν δεν τους κουβανείς μες στην ψυχή σου,
αν η ψυχή σου δεν τους στήνει εμπρός σου.

Κι αν πτωχική την βρεις, η Ιθάκη δε σε γέλασε.
Έτσι σοφός που έγινες, με τόση πείρα,
ήδη θα το κατάλαβες η Ιθάκες τι σημαίνουν.

Κωνσταντίνος Π. Καβάφης (1911)

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# Abstract

The work described in this thesis is done in the context of a long term effort at VERIMAG laboratory to build a complete model based tool-chain for the design and implementation of embedded systems. We follow a layered approach that distinguishes the application level from the architectural/implementation level. The application is described in a high-level language that is independent of implementation details. The application is then mapped to a specified architecture using a number of techniques so that the desired properties of the high level description are preserved.

In this thesis, the application is described in SIMULINK/STATEFLOW, a wide-spread modeling language that has become a de-facto standard in many industrial domains, such as automotive. At the architectural level we consider single-processor, multi-tasking software implementations. Multi-tasking means that the application software is divided into a number of processes that are scheduled by a real-time operating system, according to some preemptive scheduling policy, such as static-priority or earliest deadline first.

Between these two layers we add an intermediate representation layer, based on the synchronous language LUSTRE, developed at VERIMAG for the last 25 years. This intermediate layer permits us to take advantage of a number of tools developed for LUSTRE, such as simulators, model-checkers, test generators and code generators.

In a first part of the thesis, we study how to translate automatically SIMULINK/STATEFLOW models into LUSTRE. For SIMULINK this is mostly straightforward, however it still requires sophisticated algorithms for the inference of type and timing information. The translation of STATEFLOW is much harder for a number of reasons; first STATEFLOW presents a number of semantically "unsafe" features, including non-termination of a synchronous cycle or dependence of semantics on the graphical layout. Second, STATEFLOW is an automata-based, imperative language, whereas LUSTRE is a dataflow language. For the first set of problems we propose a number of statically verifiable conditions that define a "safe" subset of STATEFLOW. For the second set of problems we propose a set of techniques to encode automata and sequential code into dataflow equations.

In the second part of the thesis, we study the problem of implementing synchronous designs in the single-processor multi-tasking architecture described above. The crucial issue is how to implement inter-task communication so that the semantics of the synchronous design are preserved. Standard implementations, using single buffers protected by semaphores to ensure atomicity, or other, lock-free, protocols proposed in the literature do not preserve the synchronous semantics. We propose a new buffering scheme that preserves the synchronous semantics and is also lock-free. We also show that this scheme is optimal in terms of buffer usage.

# Résumé

Le travail décrit dans cette thèse fait partie d'un effort de recherche au laboratoire VERIMAG pour créer une chaîne d'outils basée sur modèles (model-based) pour la conception et l'implantation des systèmes embarquées. Nous utilisons une approche en trois couches, qui séparent le niveau d'application du niveau implantation/architecture. L'application est décrite dans un langage de haut niveau qui est indépendante des détails d'implantation. L'application est ensuite transférée à l'architecture d'exécution en utilisant des techniques spécifiques pour que les propriétés demandées soient bien préservées.

Dans cette thèse, l'application est décrite en SIMULINK/STATEFLOW, un langage de modélisation très répandu dans le milieu de l'industrie, comme celui de l'automobile. Au niveau de l'architecture, nous considérons des implantation sur une plate-forme "mono-processeur" et "multi-tâches". Multi-tâches signifie que l'application est répartie en un nombre des tâches qui sont ordonnées par un système d'exploitation temps-réel (RTOS) en fonction d'une politique d'ordonnancement préemptive comme par exemple la priorité statique (static-priority SP) ou la date-limite la plus proche en priorité (earliest deadline first EDF).

Entre ces deux couches, on rajoute une couche de représentation intermédiaire basée sur le langage de programmation synchrone Lustre, développé à VERIMAG durant les 25 dernières années. Cette représentation intermédiaire permet de profiter des nombreux outils également développés à VERIMAG tels que des simulateurs, des générateurs de tests, des outils de vérification et des générateurs de code.

Dans la première partie de cette thèse, on étudie comment réaliser une traduction automatique de modèle SIMULINK/STATEFLOW en modèles Lustre. Coté SIMULINK, le problème est relativement simple mais nécessite néanmoins l'utilisation d'algorithmes sophistiqués pour inférer correctement les informations de temps et de types (de signaux) avant de générer les variables correspondantes dans le programme Lustre. La traduction de STATEFLOW est plus difficile à cause d'un certain nombre de raisons ; d'abord STATEFLOW présent un certain nombre de comportements "non-sûr" tels que la non-terminaison d'un cycle synchrone ou des sémantiques qui dépendent de la disposition graphique des composants sur un modèle. De plus STATEFLOW est un langage impératif, tandis que Lustre un langage de flots de données. Pour le premier problème nous proposons un ensemble de conditions vérifiables statiquement servant à définir un sous-ensemble "sûr" de STATEFLOW. Pour le deuxième type de problèmes nous proposons un ensemble de techniques pour encoder des automates et du code séquentiel en équations de flots de données.

Dans la deuxième partie de la thèse, on étudie le problème de l'implantation de programmes

synchrones dans l'architecture mono-processeur et multi-tâche décrite plus haut. Ici, l'aspect le plus important est comment implanter les communications entre tâches de manière à ce que la sémantique synchrone du système soit préservée. Des implantations standards, utilisant des buffers de taille un, protégés par des sémaphores pour assurer l'atomicité, ou d'autres protocoles "lock-free" proposés dans la littérature ne préservent pas la sémantique synchrone. Nous proposons un nouveau schéma de buffers, qui préserve la sémantique synchrone tout en étant également "lock-free". Nous montrons de plus que ce schéma est optimal en terme d'utilisation des buffers.

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# Chapter 1

# Introduction

## Context of this thesis

The technological evolution we are experiencing the last decades has resulted in using more and more "computerized" devices for an entire spectrum of tasks, from the simple everyday tasks to the most sophisticated and complicated ones, which is far from the classical perception of what we used to call "computer". Moreover the constantly reducing size of these systems results to finding usage even in the most unthinkable (for the past decades) places, providing the market with a big number of small devices that we may call "gadgets". More than 98% of processors today are in such systems, and are no longer visible to the customer as computers in the classical sense.

Other examples can be found in consumer electronics such as mobile phones, house electrical appliances or in more "heavy" industries like automotive, railway, avionics and aerospace. We call all these systems *embedded systems*. More precisely, when we consider high-risk application domains, like the avionics or the automotive, those systems are called *safety-critical embedded systems*. In this specific area of embedded and real-time systems the basic requirement is high-quality design and being able to guarantee a set of correctness properties.

The design of such systems is a difficult task. In this context, the *model-based design* paradigm has been established as an important paradigm for the development of modern embedded software. The main principle of the paradigm is to use models (with formal or semi-formal semantics) all along the development cycle, from design, to analysis, to implementation. Using models rather than, say, building prototypes is essential for keeping the development costs manageable. However, models alone are not enough. Especially in the safety-critical domain, they need to be accompanied by powerful tools for analysis (e.g., model-checking) and implementation (e.g., code generation). Automation here is the key: high system complexity and short time-to-market make model reasoning a hopeless task, unless it is largely automatized.

The benefits of model-based design are many. First, high-level models raise the level of abstraction, allowing the designer to focus on the essential functions of the system rather than implementation details. This in turn makes possible to build larger and more complex systems. Analysis tools, such as simulation or model-checking tools, are crucial at this stage. Bugs that are found early in the design process are easier and less expensive to fix.

At some point the implementation phase begins, during which the system is actually built. By "system" we mean hardware, software or both. One may consider also the environment as part of the system and thus of the whole implementation process[1]. In the hardware industry the implementation phase is closely coupled with the modeling and analysis phase. Powerful EDA tools, stemming from a rich body of research on logic synthesis and similar topics, are used for gate synthesis, circuit layout, routing, etc. Such tools are largely responsible for the proliferation of electronics and their constant increase in performance.

In the software industry the situation is not as clear. On one hand, high-level models are not as widespread. After all, the software itself *is* a model and simulation can be done by executing the software. Testing and debugging are common-place (in fact, very time-consuming) but they are done at the implementation level, that is, on the target software. Implementation is automated using the most classical tools in computer science: compilers. The situation is changing, however: languages such as SIMULINK/STATEFLOW [2], UML[3] and others, as well as corresponding software-synthesis facilities are used more and more. Currently, software synthesis is mostly restricted to separate code generation for parts of the system. The integration of the different pieces of code is usually done "manually" and is source of many problems, since the implementation often exhibits unexpected behavior: deadlocks, missed data values, etc. These problems arise because the implementation method (in this case, code generation followed by manual integration) does not guarantee that the original behavior (high-level semantics) is preserved.

The reason high-level semantics are generally not preserved by straightforward implementation methods is the fact that high-level design languages often use "ideal" semantics, such as concurrent, zero-time execution (as well as communication) of an unlimited number of components. It is essential to use such ideal semantics in order to keep the level of abstraction sufficiently high. On the other hand, these assumptions break down upon implementation: components take time to execute; communication is also not free; neither is concurrency: scheduling is needed when many software components are executed on the same processor and communication is needed between components executing on separate hardware platforms.

As a result, implementations often exhibit a very different behavior than the original high-level model. This is a problem, because it means that the results obtained by analyzing the model (e.g., model satisfies a given property) may not hold at the implementation level. In turn this implies that testing and verification need to be repeated at the implementation level. This is clearly a waste of resources. In order to avoid this, we need to address the issue of the preservation of properties of the high-level model, when moving towards the implementation.

This is the vision that motivates this thesis. To contribute to the effort of building a complete model-based tool-chain that starts with high-level design languages and allows automatic, as much as possible, synthesis of embedded code that, by construction, preserves crucial properties of the high-level model.

This is an ambitious goal and, naturally, we had to look at only some parts of the entire problem. In particular, our main focus has been the class of embedded control applications, and

[1] This is especially important in control applications where the environment is the object to be controlled and the one the controller needs to be adapted to.

[2] MATLAB, SIMULINK and STATEFLOW are trademarks of the *MathWorks Inc.*: http://www.mathworks.com.

[3] From the Object Management Group: http://www.uml.org/

mostly those coming from domains such as automotive. Many of our choices, such as the choice of which high-level design languages to consider, are a result of this context.

Our work has been performed in the context of two European projects, the project "NEXT TTA"[4] and the project "RISE"[5]. We should also say that our work has been part of an on-going team effort at the VERIMAG laboratory for a number of years. This effort is explained in more detail in Chapter 2.

Before proceeding to list the contributions of this thesis, let us briefly describe the state of the art in what concerns model-based design.

## State of the art

In the domains that we are mostly interested, that is the safety-critical applications like the automotive and avionic industries, the use of the SIMULINK/STATEFLOW toolkit by *MathWorks* is considered as the *de-facto* standard.

SIMULINK offers to the designer a graphical interface that allows to combine blocks from a number of libraries and interconnect them using signals. These blocks may perform basic operations (like addition or multiplication) or more advanced ones (like transfer functions or integration). SIMULINK can model discrete as well as continuous behavior making it feasible for the designer to model not only the system but also the physical environment where this system will be embedded. Thus the designer can simulate the interaction of his system and the environment to check for the correctness and to measure the performance. Coupled with SIMULINK there is the STATEFLOW tool that provides automata-based design capabilities. Other products of *MathWorks* are the REAL-TIME WORKSHOP and REAL-TIME WORKSHOP EMBEDDED CODER code generators, that produce, starting from a SIMULINK/STATEFLOW model, imperative code for given target execution platforms. Other companies also provide third-party tools for SIMULINK/STATEFLOW, such as the TARGETLINK library and code generator from *d*Space.

SIMULINK/STATEFLOW started purely as a simulation environment and lacks many desirable features of programming languages. It has a multitude of semantics (depending on user-configurable options), informally and sometimes only partially documented. Although commercial code generators exist for SIMULINK/STATEFLOW, these present major restrictions. For example, TARGETLINK does not generate code for blocks of the "Discrete" library of SIMULINK, but only for blocks of the dSpace-provided SIMULINK library, and currently handles only mono-periodic systems. Another issue not addressed by these tools is the preservation of semantics. Indeed, the relation between the behavior of the generated code and the behavior of the simulated model is unclear. Often, speed and memory optimization are given more attention than semantic consistency. We describe in detail the short-comings of current code-generators for SIMULINK in Section 7.9.

[4] The project's title is "High-Confidence Architecture for Distributed Control Applications" and for more information refer to the official web-page http://www.vmars.tuwien.ac.at/projects/nexttta

[5] "RISE" stands for "Reliable Innovative Software for Embedded Systems" and more information can be found in http://www.esterel-technologies.com/rise/

SCADE, a product of *Esterel Technologies, Inc.*, is another design environment for embedded software. SCADE uses a graphical user interface for design capture, similar to SIMULINK. However, at the heart of the tool lies a backbone language, LUSTRE [HCRP91], which is a synchronous language with formal semantics, developed at the VERIMAG laboratory for the past twenty years. SCADE features model-checking capabilities with its "plug-in" from Prover [6], a very important feature in the domain of safety-critical applications. Moreover, SCADE is endowed with a DO178B-level-A[7] code generator which allows it to be used in highest criticality applications. SCADE has been used in important European avionic projects (Airbus A340-600, A380, Eurocopter) and is also becoming a de-facto standard in this field.

Besides these and other commercial products[8], there is a number of related offerings from the academic world. *Synchronous languages* is one set of such offerings [BCE+03]. These languages appeared at approximately the same time in the eighties and include LUSTRE [HCRP91], ESTEREL [BG92] and SIGNAL [GGBM91]. Synchronous languages share a number of common characteristics, among which a deterministic, synchronous automaton semantics, much like that of a Mealy machine. These languages were conceived early-on as high-level programming languages, targeted at embedded software. Code generation for these languages has thus been one of the main concerns, and a lot of effort has been devoted in this direction [HCRP91, BG92, GGBM91].

With a different focus, and larger scope, the Metropolis project [BWH+03] offers a framework that implements the *platform-based design* paradigm. In this paradigm, function (i.e., high-level model) and architecture (i.e., execution platform) are clearly separated. Metropolis offers a modeling framework to capture both. It also provides mechanisms, essentially by means of action synchronization, for *mapping* the function onto the architecture. This mapping (currently chosen by the user) essentially defines an implementation choice. By trying out and evaluating different mappings, the user can explore different implementation choices.

The PTOLEMY project[9] is another framework mostly focusing in modeling, and in particular in *heterogeneous* formalisms, that use radically different *models of computation*. Examples of such formalisms range from Kahn process networks [Kah74] to Communicating Sequential Processes (CSP) [Hoa85] to hybrid automata [ACH+95]. How to compose such heterogeneous models in a coherent manner is a non-trivial problem, and the main focus of PTOLEMY.

## Contributions of this thesis

State-of-the-art offerings are limited in a number of ways. Some solutions, for instance SIMULINK, lack in formal semantics and provide little analysis capabilities except simulation[10].

[6] http://www.prover.com

[7] DO-178B, Software Considerations in Airborne Systems and Equipment Certification is a standard for software development, which was developed by RTCA and EUROCAE. The FAA accepts use of DO-178B as a means of certifying software in avionics.

[8] e.g., Telelogic's TAU, supporting UML and SysML (http://www.telelogic.com/Products/tau/), ETAS' ASCET automotive platform, and more

[9] http://ptolemy.eecs.berkeley.edu [BHLM94]

[10] A recent plug-in to SIMULINK is the "Verification and Validation" tool-box which performs input stimuli generation and checks model coverage. The third-party tool Reactis, by Reactive Systems, has similar functionality.

Others, for instance synchronous languages, provide relatively restricted modeling frameworks. Finally, most solutions provide very limited code generation capabilities (essentially single-processor, single-tasking) with little or no guarantees on preservation of properties of the high-level model.

With this work, we hope to provide remedies to some of these shortcomings. Our work has been done in the context of a long-term effort at the VERIMAG laboratory aiming to build a complete tool-chain for the design and the implementation of safety-critical embedded systems (see Chapter 2 for details). The contributions that this thesis brings to this effort are the following:

- We provide a method to translate (discrete-time) SIMULINK to LUSTRE. SIMULINK offers an excellent high-level modeling framework, that is widespread in a number of application domains, in particular, in the automotive domain. Our work provides a way to link SIMULINK to a number of tools available for LUSTRE, including formal analysis tools such as model-checkers and test generators. The translation also allows to use the code generation capabilities for LUSTRE as well as the new techniques developed in this thesis (see below). This translation is presented in Chapter 4.

- We provide a method to translate STATEFLOW to LUSTRE. STATEFLOW offers enhanced modeling capabilities to SIMULINK. Together they offer a heterogeneous modeling framework based on a combination of block diagrams and state machines. Apart from our translation to LUSTRE, we also provide static analysis techniques for STATEFLOW, that permit to guarantee absence of critical errors in a model, such as non-termination of a simulation cycle. This translation is presented in Chapter 5.

- We provide tools that implement the above translation methods. We also describe a few case studies we have treated using these tools. The tools and case studies are presented in Chapter 6.

- We provide a method to generate code from synchronous models onto single-processor, multi-tasking execution platforms. Our method is proved to be semantics-preserving, in the sense that the executable software has the same behavior as the "ideal" synchronous model. Our method is also optimal with respect to memory requirements. The method is presented in Chapter 7.

## Organization of this document

The rest of this document is organized as follows. Chapter 2 describes the overall model-based design effort at VERIMAG, in the context of which this work has been performed. In Chapter 3 we provide the basics of the LUSTRE synchronous programming language, which is a crucial element in this overall approach. In Chapters 4 and 5 we study the translation of SIMULINK and STATEFLOW, respectively, to LUSTRE. In Chapter 6 we present the implementation of these translation methods in the tool SS2LUS, and also describe some case studies where this tool has been used. Finally in Chapter 8 we present the conclusions of this work and possible future directions.

# Chapter 2

# Model-based design at VERIMAG

This thesis is in the context of almost 10 years efforts in VERIMAG to build a complete model based tool-chain for the design and implementation of embedded systems. Moreover we focus in the embedded systems that are used in high-risk application domains such as the automotive and the aeronautics, where security is the most important requirement.

Designing according to the model based approach refers to using high-level abstractions to conceptualize a system. Those high-level abstractions may be software platforms or models that abstract the real implementation details and provide the designer with means to focus in certain aspects of his design. Most of the times it is better to provide the designer with a methodology that abstracts the communications between tasks of his model. Having a more general tool is redundant and complicates the design in this case.

We propose a layered approach that distinguishes the application and the architecture in which the execution will take place. In this sense, our approach follows the paradigm of *platform-based design* [SV02], implemented in other frameworks such as *Metropolis* [BWH+03].

In our flavor of this approach we add another layer between the application and the architecture layers. This layer that serves as an intermediate representation, is the LUSTRE synchronous programming language, developed in VERIMAG for the last 25 years. We choose LUSTRE first because of the vast knowledge of its mechanisms available within VERIMAG and also because it permits us to take advantage of a number of tools developed for it, such as simulators, model-checkers, test generators and code generators. A more detailed description of LUSTRE is provided in Chapter 3.

The model based approach we propose is the three-layered one we can see in Figure 2.1. As said earlier, the LUSTRE language serves as an intermediate level between the top and the bottom pyramids.

The top level is the application layer and contains all the tools, platforms and models that can be used to facilitate the design of a system. In this area we can find SIMULINK, STATEFLOW, UML or other formalisms and mathematical representations.

In the middle level we position LUSTRE, into which we translate the models produced in the top level. Then we use the model-checking capabilities of LUSTRE to verify and validate our design and thus, be sure that we the system respects always some safety properties. The importance of the latter is significant since, as we saw earlier, the target applications are the ones

in the domains of automotive, aerospace and more general in high-risk applications.

The bottom level is the one of the actual execution of the system. In this layer we have all the possible execution architectures and we can choose according to our needs. This contains centralized or distributed architectures using single-tasking or multi-tasking software, various hardware platforms as well as various communication media.

An on-going effort at VERIMAG aims at enriching the top and bottom layers, adding more high-level description languages and more low-level execution platforms, to the capabilities of our current tools. The first compiler of the LUSTRE language [HCRP91] covers implementation of LUSTRE on a *mono-processor* and *single-task* execution platform. The work of Adrian Curic [Cur05, CCM+03] covers implementation of LUSTRE (and an extended LUSTRE language) to a distributed *synchronous* execution platform, called the *Time-Triggered Architecture* (TTA) [Kop97], where nodes communicate via the *Time-Triggered Protocol* (TTP) [KG94].

Our work contributes to the previous efforts as depicted by the dashed lines in Figure 2.1. First, we provide a translation from SIMULINK to LUSTRE that we study in Chapter 4. We also study the translation of STATEFLOW to LUSTRE in Chapter 5. Finally, we study the implementation of LUSTRE (and synchronous languages in general) on a mono-processor, *multi-tasking* execution platform in Chapter 7.

We hope that future works will further enhance the picture by considering other high-level languages (e.g., UML [Gro01]) and more execution platforms.

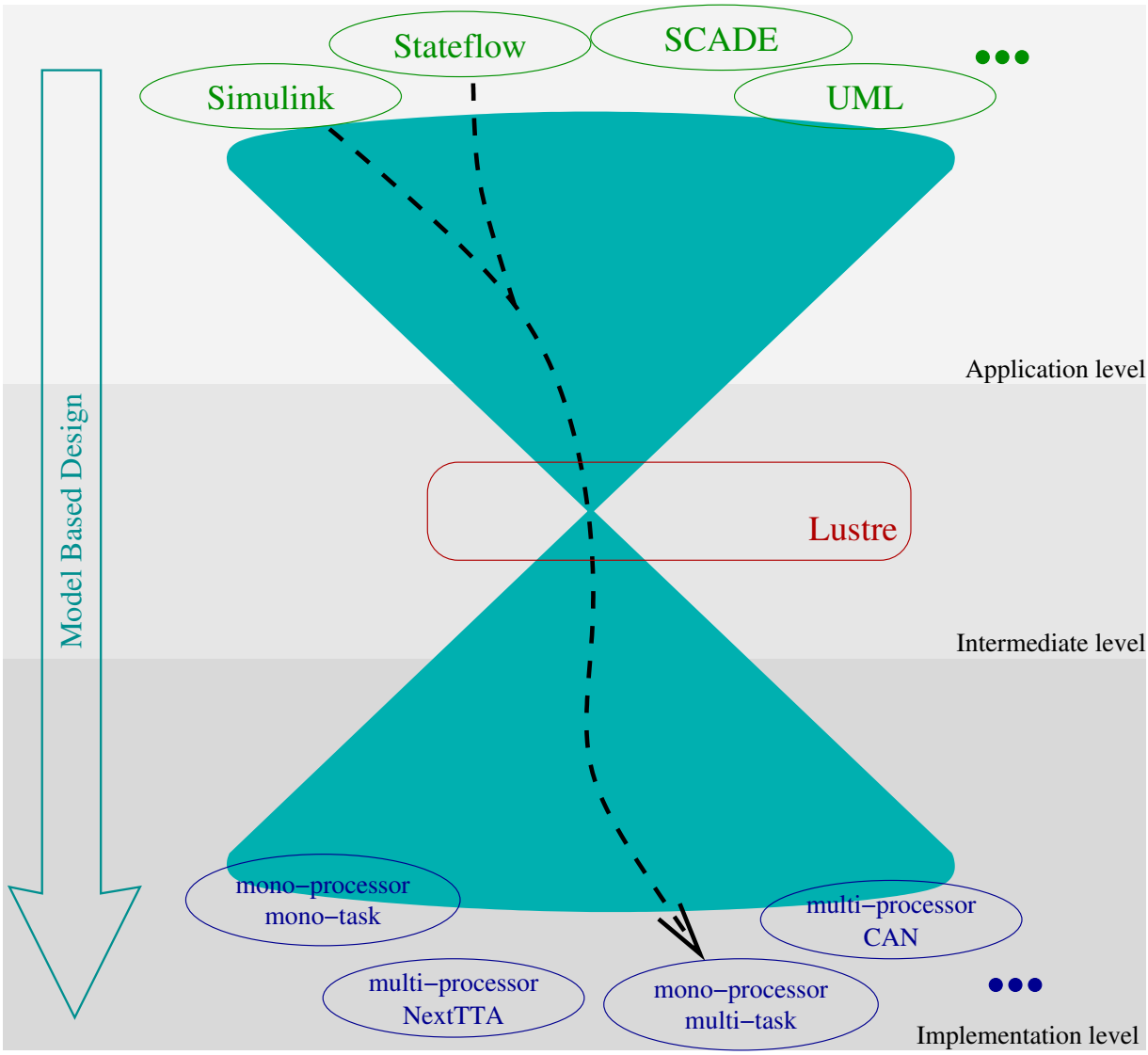



**Figure 2.1:** Model Based Design architecture.

# Chapter 3

# The Synchronous Programming Language LUSTRE

LUSTRE [CPHP87, HCRP91] is a synchronous dataflow programming language that has been designed and developed in VERIMAG laboratory 25 years ago, following the merging of the synchronous and the dataflow paradigms. On top of the language there is a number of tools that constitute the LUSTRE platform. These tools take advantage of the formal foundations of LUSTRE to provide model-checking capabilities and test generation. There is also the compiler of producing imperative C code, that respects the semantics of the language. Finally there are tools for the simulation of the system on design.

Moreover, SCADE, the industrial counterpart of LUSTRE, was founded by *Esterel Technologies, Inc.*. SCADE, having LUSTRE as backbone, provides graphical user interface to the designer and also has a *certified* compiler[1], which is very important in the area of aeronautics and automotive industries. We can measure the importance of SCADE, and thus of LUSTRE, by counting the number of companies already using it as a basis for their design. The last big success of SCADE is its use for the development of the latest project of Airbus, the A380 carrier airplane.

In this Chapter, we demonstrate the principles of LUSTRE and also give some examples of usage. Furthermore, we discuss the C code generation capabilities of the LUSTRE compiler. We do not intent to provide a full-fledged cover of the area, in which case the reader should refer to [BCGH93, BCE+03].

## 3.1 The LUSTRE language

A LUSTRE program models essentially a deterministic Mealy machine, as illustrated in Figure 3.1. The machine has a (possibly infinite) set of states, a (possibly infinite) set of inputs and a (possibly infinite) set of outputs. Given the current state and the current input the transition function and output function compute the next state and the current output, respectively.

[1] certified with the *DO-178B* certification on *level A*

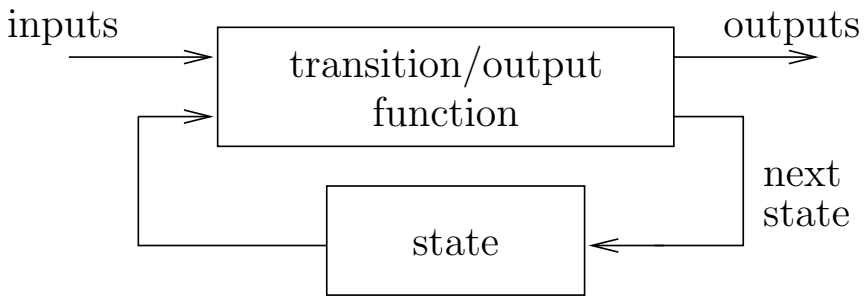



**Figure 3.1:** A LUSTRE program is a deterministic Mealy machine.

LUSTRE time is *logical* and counted in discrete instants. Logical time means there is no a-priori notion of instant duration or time elapsing between two instants. Indeed, the beginning of each instant is determined at run-time by the environment in which the embedded LUSTRE program is running. This environment calls the LUSTRE program and the LUSTRE program performs one step of the Mealy machine described above: read inputs, compute outputs, write outputs. The *synchrony hypothesis* dictates that the computations associated with the current instant (reading inputs, computing outputs, writing outputs) are finished before the next instant arrives.

In LUSTRE, every variable and expression denotes a *flow*, i.e., a pair of

- a possible infinite sequence of values of a given type,
- a *clock*, representing a sequence of times.

The use of the clock is to designate the instants that the flow takes a value. This means that in the $n$-th time of its clock the flow takes the $n$-th value of its sequence of values. Any program has a cyclic behavior (the Mealy automaton discussed above) and that cycle determines the *basic clock* of the program. Other, slower clocks, can be defined using boolean-valued flows: the new clock defined by a boolean flow is the sequence of times that this flow takes the value *true*. For instance Table 3.1 displays the time-scales defined by a flow C whose clock is the basic clock and a flow C' whose clock is defined by C.

| basic time-scale (defined by the environment) | 1 | 2 | 3 | 4 | 5 | 6 | 7 | 8 |
|---|---|---|---|---|---|---|---|---|
| C | true | false | true | true | false | true | false | true |
| time-scale defined by C | 1 | | 2 | 3 | | 4 | | 5 |
| C' | false | | true | false | | true | | true |
| time-scale defined by C' | | | 1 | | | 2 | | 3 |

**Table 3.1:** Boolean flows and their clocks.

Variables in LUSTRE are coupled explicitly with their type and clock. When the latter does not appear explicitly, the clock of the variable is the basic clock of the current node, where the variable is declared. Basic data types in LUSTRE are *boolean*, *integer*, *real* and one type constructor *tuple*. Usual operators over basic types are available:

- arithmetic: **+, -, *, /, div, mod**
- boolean: **and, or, not**
- relational: **=,** $<$**,** $<=$**,** $>$**,** $>=$
- conditional: **if then else**

These are the data operators and operate **only** on operands that have the same clock; they operate point-wise on the sequences of values of their operands. For instance, if $X$ and $Y$ are two integer flows on the basic clock and their values are $(x_1, x_2, ..., x_n, ...)$ and $(y_1, y_2, ..., y_n, ...)$, then the following expression

```
if X>0 then Y+1 else 0;
```

is a flow with integer type and on the basic clock, which in the $n$-th instance the value is:
if $x_n > 0$ then $y_n + 1$ else $0$

Besides these operators, LUSTRE has four more which are called "temporal" operators, which operate specifically on flows:

- **pre** (as for "previous") acts as a memory; if $(e_1, e_2, ..., e_n, ...)$ is the sequences of values of the expression `E`, then the expression `pre E` has the same clock and its sequence of values is $(nil, e_1, e_2, ..., e_n, ...)$, where $nil$ represents an undefined value denoting an uninitialized memory.
- **->** ("followed by"): if `E` and `F` are two flows with the same clock and respectively the following values $(e_1, e_2, ..., e_n, ...)$ and $(f_1, f_2, ..., f_n, ...)$, then the expression `F ->E` is an expression with the same clock and with sequence of values $(f_1, e_1, e_2, ..., e_n, ...)$. In other words, this expression is always equal to `E` except the first instant that is equal to `F`.

Those two operators are mostly used in couple, to generate a memory with an initial value. A very "classical" way to use those operands are in combination like in the following expression:

```
Y = X -> pre Z;
```

and the result of this operation is shown in Table 3.2.

The other two temporal operators are the ones that affect the clock of a flow:

- **when** operator is used to "sample" an expression to a slower clock. Let `E` be an expression and `B` a boolean expression with the same clock, then `E when B` is a flow whose clock is defined by `B` and the sequence of values is composed by the values of `E` when `B` is `true`.

| | | | | | | | |
|---|---|---|---|---|---|---|---|
| X | $x_1$ | $x_2$ | $x_3$ | $x_4$ | $x_5$ | $x_6$ | ... |
| Z | $z_1$ | $z_2$ | $z_3$ | $z_4$ | $z_5$ | $z_6$ | ... |
| Y = X -> pre Z | $x_1$ | $z_1$ | $z_2$ | $z_3$ | $z_4$ | $z_5$ | ... |

**Table 3.2:** Example for the use of `pre` and `->`.

- **current** operator is used to "interpolate" an expression on the clock immediately faster than its own. If `E` is an expression with clock defined by the boolean flow `B`, which is not the basic clock, then `current E` has the same clock as `B` and its value is the value of `E` at the last time that `B` was `true`. Note that until the first time that `B` is `true` the resulting flow will be `nil`.

Table 3.3 provides an example to illustrate the use of the last two temporal operants.

| | | | | | | | | | |
|---|---|---|---|---|---|---|---|---|---|
| B | false | false | true | false | true | false | false | true | true |
| X | $x_1$ | $x_2$ | $x_3$ | $x_4$ | $x_5$ | $x_6$ | $x_7$ | $x_8$ | $x_9$ |
| Y = X when B | | | $x_3$ | | $x_5$ | | | $x_8$ | $x_9$ |
| current Y | $nil$ | $nil$ | $x_3$ | $x_3$ | $x_5$ | $x_5$ | $x_5$ | $x_8$ | $x_9$ |

**Table 3.3:** Example of use of `when` and `current` .

LUSTRE is a declarative language where variables are defined by equations of the form $x = E$, where $x$ is a variable and $E$ is an expression on variables or constants using the operators described above. Each intermediate or output variable in a LUSTRE program has a unique such definition, i.e., having two or more equations $x = E_1$ and $x = E_2$ is not allowed. Inputs are not defined by equations, since they are provided by the environment. An equation in LUSTRE expresses a global invariant, i.e., the value of the flow $x$ is at every logical instant equal to the value of the flow computed by $E$. Thus, equations are essentially a mechanism for functional composition.

Structure is given to a LUSTRE program by declaring and calling LUSTRE *nodes*, in much the same way as, say, C functions are declared and called. Here is an example:

`A` is a node taking as inputs a boolean flow `b`, an integer flow `i` and a real flow `x` and returning a real flow `y`. `A` uses internal flows `j` and `z` (with usual scope rules). The body of `A` is declared between the `let` and `tel` keywords. `A` calls node `B` to compute `z` and node `C` to compute `y` (conditionally). Nodes `B` and `C` are declared elsewhere.

## 3.2 LUSTRE compiler and code generation

The LUSTRE compiler guarantees that the system under design is deterministic and respects the synchronous hypothesis. It accomplishes the task thanks to static verifications which amounts to:

```
node A(b: bool; i: int; x: real)
returns (y: real);
var j: int; z: real;
let
    j = if b then 0 else i;
    z = B(j, x);
    y = 0.0 -> (if b then pre z else C(z));
tel
```

**Figure 3.2:** A LUSTRE node example.

- Definition checking: every local and output variable should have one and only one equational definition.
- Clock consistency.
- Absence of cycles in definitions: Any cycle should use at least one **pre** operator.

This latter check, in LUSTRE, is done by statically rejecting any program that contains a cycle in the instantaneous dependencies relation. This, syntactic, method of rejecting designs is some times too restrictive, since a further boolean causality check could prove that the program has one and only one solution for any given input.

Moreover to the above static checks, the LUSTRE compiler can generate code to be executed in a target platform. The first such implementation is the code generation for a mono-processor and mono-thread implementation. Thus, the compiler can generate a monolithic program in the imperative C language. The principle to associate an imperative program to LUSTRE, is to construct an infinite loop whose body implements the inputs to outputs transformations performed at any cycle of the node.

The two basic steps are: (1) introduce variables for implementing the memory needed by the `pre` operators and (2) sort the equations in order to respect data-dependencies. Note that a suitable order exists as soon as the program has been accepted by the causality checking.

```
node counter(x, reset: bool) returns (c: int);
var lc : int;
let
   c = lc + (if x then 1 else 0);
   lc = if (true -> pre reset) then 0 else pre c;
tel
```

**Figure 3.3:** A counter in LUSTRE.

Consider the example of Figure 3.3. This LUSTRE program, implements a counter that counts the occurrences of its input `x` since the last occurrence of `reset`, or the beginning of time if `reset` has never occurred.

The resulted C code generated for the example of the counter is shown in Figure 3.4, where the variables that implement the memory for the `pre` operators are `pre_reset` and `pre_c` and as for the data-dependency `lc` must be computed before `c`.

```
bool pre_reset = true;
int pre_c = 0, c; // c: output
bool x, reset;    // inputs

void counter_step() {
   int lc;
   lc = (pre_reset)? 0 : pre_c;
   c = lc + (x)? 1 : 0;
   pre_c = c;
   pre_reset = reset;
}
```

**Figure 3.4:** The C code for the example of the counter.

Following those principles, the target code is a simple sequence of assignments. The main advantage of this somehow naive algorithm is that it produces a code which is neither better nor worse than the source code: both the size and the execution time are linear with respect to the size of the source code. This one-to-one correspondence between source and target code is particularly appreciated in critical domain like avionics, and it has been adopted by the SCADE compiler.

# Chapter 4

# Analysis and Translation of Discrete-Time SIMULINK to LUSTRE

## 4.1 SIMULINK Translation Objectives

SIMULINK and LUSTRE are languages manipulating *signals* and *systems*. Signals are functions of time. Systems take as input signals and produce as output other signals. In SIMULINK, which has a graphical language, the signals are the "wires" connecting the various blocks. In LUSTRE, the signals are the program variables, called *flows*. In SIMULINK, the systems are the built-in blocks (e.g., adders, gains, transfer functions) as well as composite blocks, called *subsystems*. In LUSTRE, the systems are the various built-in operators as well as user-defined operators called *nodes*.

In the sequel, we use the following terminology. We use the term *block* for a basic SIMULINK block (e.g., adder, gain, transfer function) and the term *subsystem* for a composite SIMULINK block. We will use the term *system* for the root subsystem. We use the term *operator* for a basic LUSTRE operator and the term *node* for a LUSTRE node.

## 4.2 Differences of SIMULINK and LUSTRE

We will try now to elaborate on the differences of SIMULINK and LUSTRE languages as a first step towards the translation from the former to the latter. They are both data-flow programming languages that allow the representation of multi-periodic sampled systems as well as discrete

| | Simulink | Lustre |
|---|---|---|
| Signals | "wires" connecting blocks | variables (*flows*) |
| Systems | `Sum`, `Gain`, `Unit Delay`, ..., subsystems | `+`, `pre`, `when`, `current`, ..., nodes |

**Table 4.1:** Signals and systems in SIMULINK and LUSTRE.

event systems. However, despite their similarities, they also differ in many ways:

- LUSTRE has a discrete-time semantics, whereas SIMULINK has a continuous-time semantics[1]. It is important to note that even the blocks of SIMULINK belonging to the "*Discrete library*" produce piecewise-constant continuous-time signals. Thus, in general, it is possible to feed the output of a continuous-time block into the input of a discrete-time block and vice-versa.

- LUSTRE has a unique, precise semantics. The semantics of SIMULINK depends on the choice of a simulation method. For instance, some models are accepted if one chooses a variable-step integration solver and rejected with a fixed-step solver.

- LUSTRE is a strongly-typed language with explicit types for each flow. Explicit types are not mandatory in SIMULINK. However, they can be set using, for instance, the *Data Type Converter* block or an expression such as *single(1.2)* which specifies the constant 1.2 having type *single*. The differences of the typing mechanisms of SIMULINK and LUSTRE are discussed in more detail in Section 4.4.

- LUSTRE is modular in certain aspects, while SIMULINK is not. In particular, concerning timing aspects, a SIMULINK model allows a subsystem to "run faster" (be sampled at a higher rate) than its parent system. In this sense, SIMULINK is not modular since the subsystem contains implicit inputs (i.e., sample times). The differences of the timing mechanisms of SIMULINK and LUSTRE are discussed in more detail in Section 4.5.2.

- Hierarchy in SIMULINK is present both at the definition and at the execution levels. This means that subsystems are drawn graphically within their parent systems, to form a tree-like hierarchy. The same hierarchy is used to determine the order of execution of nodes. In LUSTRE, the execution graph is hierarchical (nodes calling other nodes), while the definition of nodes is "flat" (that is, following the style of C rather than, say, Pascal, where procedures can be declared inside other procedures). The differences of the structure of SIMULINK and LUSTRE are discussed in more detail in Section 4.6.

## 4.3 The Goals and Limitations of the Translation

The ultimate objective of the translation is to automatize the implementation of embedded controllers as much as possible. We envisage a tool chain where controllers are designed in SIMULINK, translated to LUSTRE, and implemented on a given platform using the LUSTRE C code generator and a C compiler for this platform. Other tools, for instance, for worst-case execution time (WCET) analysis, code distribution, schedulability analysis and scheduling, etc., can also assist the implementation process, especially when targeting distributed execution platforms (e.g., see [CCM+03]).

[1] In the sense that LUSTRE signals are functions from the set of natural numbers to sets of values and SIMULINK signals are functions from the set of positive real numbers to sets of values.

The basic assumption is that the embedded controller to be implemented is designed in SIMULINK using the *discrete-time* part of the model. Thus, we only translate the discrete-time part of a SIMULINK model. Of course, controllers can be modeled in continuous time as well. This is typically done in control theory, so that analytic results for the closed-loop system can be obtained (e.g., regarding its stability). Analytical results can also be obtained using the sampled-data control theory. In any case, the implemented controller must be discrete-time. How to obtain this controller is a control problem which is clearly beyond the scope of this paper. According to classical text-books [AW84], there are two main ways of performing this task: either design a continuous-time controller and sample it or sample the environment and design a sampled controller.

Concretely, the subset of SIMULINK we translate includes blocks of the "*Discrete*" library such as "Unit-delay", "Zero-order hold", "Discrete filter" and "Discrete transfer function", generic mathematical operators such as sum, gain, logical and relational operators, other useful operators such as switches, and, finally, subsystems including triggered and enabled subsystems. The subset of SIMULINK currently handled by our method and tool (called S2L) is shown in Figure 4.1.

Other goals and limitations of our translation are the following.

1. We aim at a translation method that preserves the semantics of SIMULINK. This means that the original SIMULINK model and the generated LUSTRE program should have the same observable output behavior, given same inputs, modulo precisely defined conditions. Since SIMULINK semantics depends on the simulation method, we restrain ourselves only to one method, namely, "*solver: fixed-step, discrete*" and "*mode: auto*". We also assume that the LUSTRE program is run at the time period the SIMULINK model was simulated. Thus, an outcome of the translation must be the period at which the LUSTRE program *shall* be run (see also Section 4.5).

2. We do not translate *S-functions* or *Matlab functions*. Although such functions can be helpful, they can also create side-effects, which is something to be avoided and contrary to the "functional programming" spirit of LUSTRE. Notice, however, that our tool does not "block" or rejects the input model when the latter contains such functions. It translates them into *external function* calls, like other "unknown" SIMULINK blocks (see also item 5, below).

3. As the SIMULINK models to be translated are in principle controllers embedded in larger models containing both discrete and continuous parts, we assume that for every input of the model to be translated (i.e., every input of the controller) the sampling time is explicitly specified. This also helps the user to see the boundaries of the discrete and the continuous parts in the original model.

4. In accordance with the first goal, we want the LUSTRE program produced by the translator to type-check if and only if the original SIMULINK model type-checks (i.e., is not rejected by SIMULINK because of type errors). However, the behavior of the type checking mechanism of SIMULINK depends on the simulation method and the "*Boolean logic signals*"

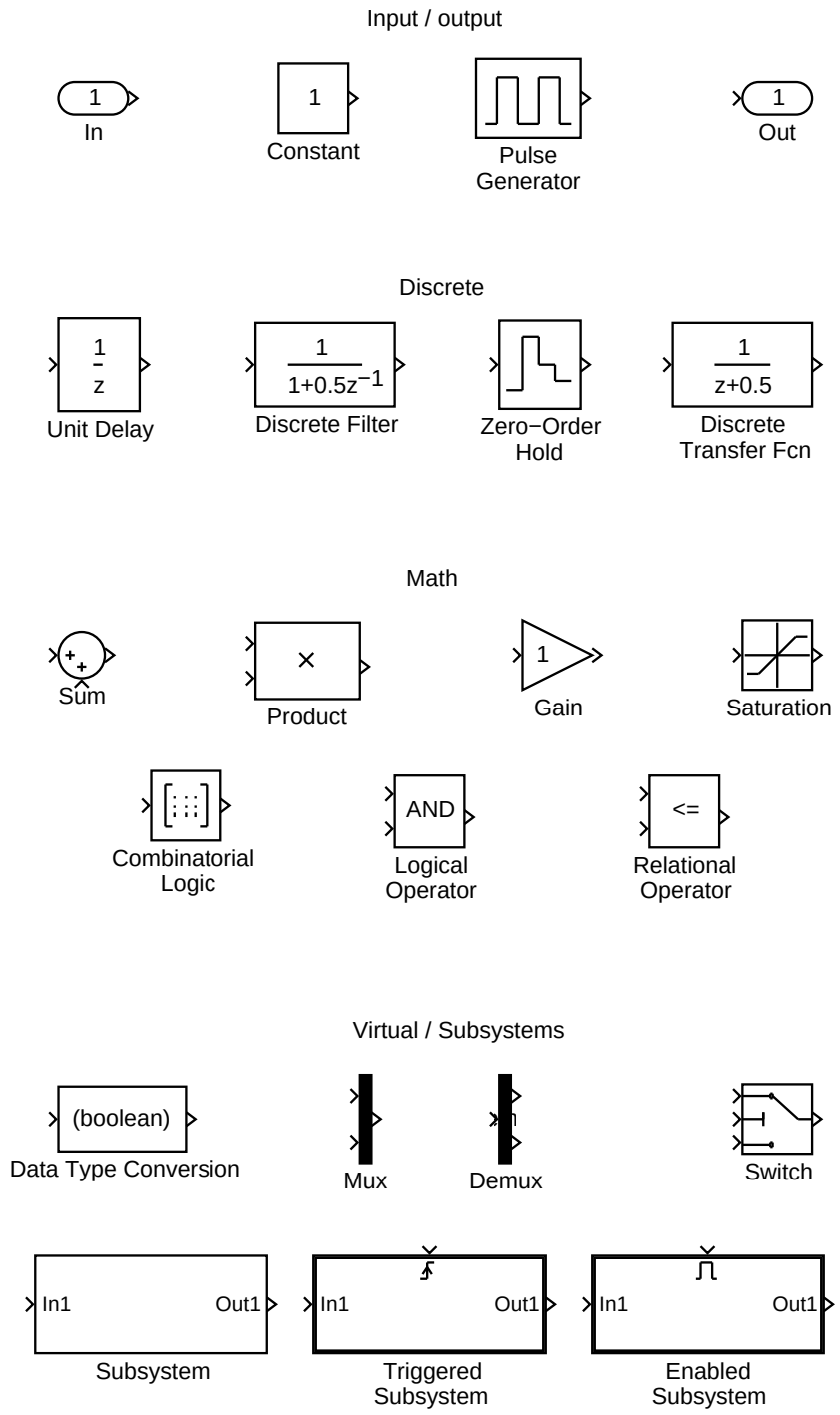



**Figure 4.1:** The set of SIMULINK blocks that are currently handled by S2L.

flag (BLS). Thus, apart from the simulation method which must be set as stated in item 1, we also assume that BLS is on. When set, BLS imposes that inputs and outputs of logical blocks (and, or, not) be of type *boolean*. Not only this is good modeling and programming practice, it also makes type inference more precise (see also Section 4.4) and simplifies the verification of the translated SIMULINK models using LUSTRE-based model-checking tools.

We also set the "*algebraic loop*" detection mechanism of SIMULINK to the strictest degree, which rejects models with such loops. These loops correspond to cyclic data dependencies in the same instant in LUSTRE. The LUSTRE compiler rejects such programs.

5. For reasons of traceability, the translation must preserve the hierarchy of the SIMULINK model as much as possible. We achieve this by suitable naming, as explained in Section 4.6.

6. The translator must "try its best". This means that it must be able to handle as larger a part of the SIMULINK model as possible, potentially leaving some parts un-translated. But it should not "block" or reject models simply because there are some parts in them that the translator does not "know" how to translate. We achieve this by taking advantage of the possibility to include *external data types* and *external functions* in a LUSTRE program. The translator generates external function code for every "unknown" block.

It should also be noted that SIMULINK is a product evolving in time. This evolution has an impact on the semantics of the tool, as mentioned earlier. For instance, earlier versions of SIMULINK had weaker type-checking rules. We have developed and tested our translation method and tool with Matlab version 6.5.0 release 13 (Jun 18, 2002), SIMULINK Block Library 5.0.1. All examples given in the thesis refer to this version as well. However any SIMULINK model created with a MATLAB release between r12 and r13 is treated and translated correctly.

## 4.4 Type inference

Type inference is a prior step to translation per se. It is necessary in order to infer the types of signals in the SIMULINK model and use them to associate types of variables in the generated LUSTRE program. In this section, we explain the typing mechanisms of LUSTRE and SIMULINK and then present the type inference technique we use. The type rules for SIMULINK that are stated here are with respect to the simulation method and flag options mentioned in Section 4.3.

### 4.4.1 Types in LUSTRE

LUSTRE is a strongly typed language, meaning that every variable has a declared type and operations have precise type signatures. For instance, we cannot add an integer with a boolean or even an integer with a real. However, predefined *casting* operators such as `int2real` can be used to transform the type of a variable in a "safe" way.

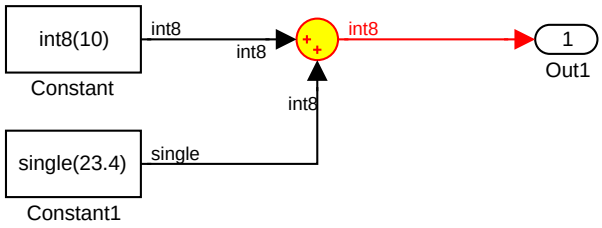



**Figure 4.2:** A type error in SIMULINK.

Basic types in LUSTRE are `bool` for boolean, `int` for integer and `real`. Composite types are essentially *records* of fixed size, constructed with the `type` operator. For instance,

```
type type_rrb = {real, real, bool};
```

declares the new type `type_rrb` as the tuple of two reals and one boolean. The LUSTRE compiler ensures that operations between flows with different types do not take place. For example,

```
var x: int; y: real; z: real;

z = x + y;
```

results into a type error and the program is rejected.

It should be noted that constants in LUSTRE also have types. Thus, `0` is zero of type `int`, whereas `0.0` is zero of type `real`. `true` and `false` are constants of type `bool`.

### 4.4.2 Types in SIMULINK

In SIMULINK, types need not be explicitly declared. Nevertheless, SIMULINK does have typing rules: some models are rejected because of type errors. An example is shown in Figure 4.2. The model contains a type error since a signal of type `int8` and a signal of type `single` are attempted to be fed as inputs to the `Sum` block which expects its inputs to have the same type. Notice that the SIMULINK simulator detects this error. The annotations of signals with types and colors in the figure is performed by SIMULINK itself.

The objective of the type inference step is to find the type of each SIMULINK signal. This type is then mapped during the translation step to a type of the corresponding variable in the generated LUSTRE program.

Informally, the type system of SIMULINK can be described as follows. There are 9 basic "*data types*", namely, `boolean`, `double`, `single`, `int8`, `uint8`, `int16`, `uint16`, `int32` and `uint32`. By default, all signals are `double`, except when:

1. the user explicitly sets the type of a signal to another type, e.g., by a `Data Type Converter` block or by an expression such as `single(23.4)`; or

2. a signal is used in a block which demands another type. For instance, all inputs and outputs of `Logical Operator` blocks are required to be `boolean`.

| | | |
|---|---|---|
| `Constant`$_\alpha$ | : | $\alpha, \alpha \in SimNum$ |
| `Sum` | : | $\alpha \times \cdots \times \alpha \rightarrow \alpha, \alpha \in SimNum$ |
| `Gain` | : | $\alpha \rightarrow \alpha, \alpha \in SimNum$ |
| `Relation` | : | $\alpha \times \alpha \rightarrow$ `boolean`$, \alpha \in SimNum$ |
| `Switch` | : | $\alpha \times \beta \times \alpha \rightarrow \alpha, \alpha, \beta \in SimNum$ |
| `Logical Operator` | : | `boolean` $\times \cdots \times$ `boolean` $\rightarrow$ `boolean` |
| `Discrete Transfer Function` | : | `double` $\rightarrow$ `double` |
| `Zero-Order Hold, Unit Delay` | : | $\alpha \rightarrow \alpha, \alpha \in SimT$ |
| `Data Type Converter`$_\alpha$ | : | $\beta \rightarrow \alpha, \alpha, \beta \in SimT$ |
| `InPort, OutPort` | : | $\alpha \rightarrow \alpha, \alpha \in SimT$ |

**Table 4.2:** Types of some SIMULINK blocks.

A type error occurs when incompatible types are fed into a block, for instance, when a `boolean` is fed into a `Sum` block or when an `int8` is fed into a `Discrete Transfer Function` block.

Denote by $SimT$ the set of all SIMULINK types and let $SimNum = SimT - \{\texttt{boolean}\}$. Then, the typing requirements imposed by SIMULINK basic blocks can be represented by the *type signatures* given in Table 4.2. It can be seen that some blocks impose no typing constraints, such as the `Unit Delay` block, while others have precise type signatures, such as the `Logical Operator` blocks (logical and, or, etc.).

Thus, the type system of SIMULINK is *polymorphic* and in fact contains both types of polymorphism (e.g., see [Pie02]):

- *Parametric polymorphism*: this is, for instance, the case of the `Unit Delay` block, which delays its input by one cycle (i.e., simulation step). The type of this block is $\alpha \rightarrow \alpha$ with $\alpha \in SimT$.

- *Ad-hoc polymorphism* (or *overloading*): this is, for instance, the situation with the `Sum` block, which accepts a number of inputs of the same numerical type and generates an output of the same type. The type of this block is $\alpha \times \cdots \times \alpha \rightarrow \alpha$ with $\alpha \in SimNum$.

### 4.4.3 Type Inference and Translation

The type system of SIMULINK can be formalized as an extension of Milner's polymorphic type system for ML [Mil78] which handles parametric polymorphism with type *classes* as used for instance in Haskell [HJW90] to handle ad-hoc polymorphism. The situation will be simpler for SIMULINK, which has only three classes, namely, $SimT$, $SimNum$ and $\{\texttt{boolean}\}$. A *unification* algorithm (e.g. [Rob65, MM82]) can be used to infer types, which is standard type theory.

For the type inference, we extract from the SIMULINK model the type equations for every *signal transformation* , using "fresh" type variables for every signal. As signal transformation, we mean all the blocks and subsystems, even though the ones that have no inputs or no outputs. After extracting all those equations, we solve them using Robinson's Unification algorithm $\mathcal{U}$.

The type of a SIMULINK subsystem (or root system) $A$ is defined given the types of the subsystems or blocks composing $A$, using a standard function composition rule.

As we shall see in Section 4.6 our translation preserves the structure of SIMULINK. In particular, every SIMULINK signal $s$ is mapped to a variable $x_s$ in the generated LUSTRE program. Once the type of $s$ is inferred using the above method, the type of $x_s$ is determined. This is done as follows.

- If the type of $s$ is `boolean` then the type of $x_s$ is `bool`.
- If the type of $s$ is `int8`, `uint8`, `int16`, `uint16`, `int32` or `uint32` then the type of $x_s$ is `int`.
- If the type of $s$ is `single` or `double` then the type of $x_s$ is `real`.
- If the type of $s$ is $\alpha$ then the type of $x_s$ is `real`.

The last case is consistent with the fact that the default type in SIMULINK is `double`.

## 4.5 Clock inference

As with type inference, clock inference is a prior step to translation. It is necessary in order to infer timing information of the SIMULINK model and use this information during generation of the LUSTRE program. In this section, we explain the timing mechanisms of LUSTRE and SIMULINK and then present our clock inference technique. Notice that the timing rules for SIMULINK that are stated

here are with respect to the simulation method and flag options mentioned in Section 4.3.

### 4.5.1 Time in LUSTRE

As mentioned in Chapter 3, LUSTRE time is logical and counted in discrete instants. Associated with each LUSTRE flow $x$ is a Boolean flow $b_x$, called the *clock* of $x$, specifying the instants when $x$ is defined: $x$ is defined at instant $i$ iff $b_x(i) = \texttt{true}$. For example, if $x$ is defined at $i = 0, 2, 4, ...$ then $b_x = \texttt{true false true false} \cdots$. We say that "$x$ runs on clock $b_x$".

Input variables are by definition defined at every instant: their clock is called the *basic* clock, represented by the Boolean flow `true`. "Slower" clocks are obtained from the basic clock using the `when` operator. For example, if `x` is an input then the flow `y` defined only at even instants can be generated by the following LUSTRE code:

```
cl_half = true -> not pre(cl_half) ;
y       = x when cl_half ;
```

| expression $e$ | clock($e$) | constraints |
|---|---|---|
| input x | *basic* | |
| x+y | clock(x) | clock(x) = clock(y) |
| pre(x) | clock(x) | |
| x when b | b | clock(x) = clock(b), b boolean |
| current(x) | clock(clock(x)) | |

**Table 4.3:** Clock calculus of LUSTRE.

cl_half is the Boolean flow alternating between true and false (starting with true). It thus defines a clock which is twice as slow as the basic clock. The expression x when cl_half defines the flow sampled from x according to clock cl_half.

Clocks can be seen as extra typing information. The compiler ensures that the LUSTRE program satisfies a set of constraints on clocks, otherwise the program is rejected. For example, in the expression x+y, x and y must have the same clock, which is also the clock of the resulting flow. The set of these constraints and how to calculate clocks is called *clock calculus*. A simplified version of this calculus is shown in Table 4.3. For more details the reader is referred to [Cas92, CP03].

It is worth mentioning that the LUSTRE compiler checks clock correctness in a *syntactic*, not *semantic* manner. Indeed, finding whether two Boolean flows are semantically equivalent is an undecidable problem[2]. Therefore, in an expression such as (x when b) + (y when c),b and c must be identical in order for the expression to clock-check.

### 4.5.2 Time in SIMULINK

SIMULINK has essentially two timing mechanisms, *sample times* and *triggers*. We briefly describe these mechanisms in what follows.

**Sample times** As mentioned already, discrete-time SIMULINK signals are in fact piecewise-constant continuous-time signals. These signals can have associated timing information, called "*sample time*" and consisting of a *period* and an initial *phase*. Sample times may be set in blocks such as input ports, unit-delay, zero-order hold or discrete transfer functions. The sample time of a signal is derived from the block producing the signal and specifies when the signal is to be updated. A signal $x$ with period $\pi$ and initial phase $\theta$ is updated only at times $k\pi + \theta$, for $k = 0, 1, 2, ...$, that is, it remains constant during the intervals $[k\pi + \theta, (k+1)\pi + \theta)$. SIMULINK requires that $\pi \geq \theta$. By default, blocks have their sample time value set to $-1$, which corresponds to an undefined or "*inherited*" (from the parent system or from the inputs) value.

Similarly to what happens to clocks in LUSTRE, sample times serve as an extra type system in SIMULINK: some models are rejected because of timing errors. An interesting example is shown

[2] This is because a Boolean flow can be seen as the output of a Turing machine. LUSTRE is Turing equivalent thus checking equivalence of Boolean flows would imply checking equivalence of Turing machines.

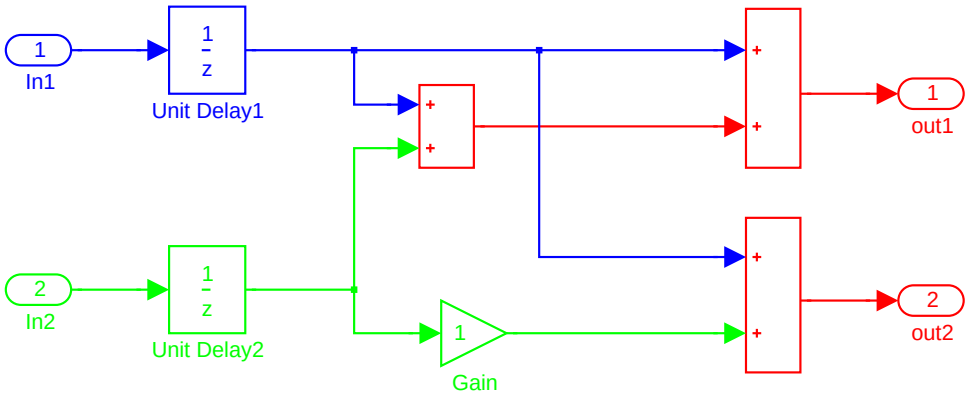



**Figure 4.3:** A SIMULINK model producing a strange error.

in Figure 4.3. The sample times of inputs "In1" and "In2" are set to 2 and 3, respectively.[3] This model is rejected by SIMULINK. However, if the `Gain` block is removed, then the model is accepted. This is strange, since the `Gain` block is a simple multiplication by 1, thus, should not be expected to change the behavior of the system. The explanation is given at the end of this section.[4]

**The "GCD rule"** Contrary to LUSTRE, SIMULINK does not require that all inputs of a block have the same sample time. The basic rule for combining signals with different sample times is what may be called the "GCD (greatest common divisor) rule". This rule states that the output of a block will have as sample time the GCD of the sample times of the inputs. An example is shown in Figure 4.4. The sample times of the two inputs of the `Sum` block are 4 and 9. Following the rule, the sample time of the output will be 1.

In fact, the rule is more complicated because sample times are not simply periods but pairs (period, phase). In general, given $n$ input signals with sample times $(\pi_i, \theta_i)$, where $\pi_i$ is the period and $\theta_i$ is the phase, for $i = 1, ..., n$, the output signal will have sample time $(\pi, \theta) = \mathsf{gcd\text{-}rule}((\pi_i, \theta_i)_{i=1,...,n})$, where

$$\begin{aligned} \pi &= \begin{cases} gcd(\pi_1, ..., \pi_n), & \text{if } \theta_1 = \cdots = \theta_n \\ gcd(\pi_1, ..., \pi_n, \theta_1, ..., \theta_n) & \text{otherwise} \end{cases} \\ \theta &= \begin{cases} \theta_1 \; mod \; \pi, & \text{if } \theta_1 = \cdots = \theta_n \\ 0, & \text{otherwise} \end{cases} \end{aligned} \tag{4.1}$$

In the above definition, $gcd$ denotes the GCD function and $mod$ the *modulo* function. For exam-

[3] Unless otherwise mentioned, we assume that phases are 0.

[4] The same anomaly persists when using Matlab Version 6.5.0.180913a Release 13 and SIMULINK version 5.0.1 (R13) dated 19-Sep-2002. But it only happens when setting simulation parameters to "fixed step", "auto".

ple,

$$\text{gcd-rule}((2,0),(3,0)) = (1,0) \qquad \text{gcd-rule}((12,3),(6,3)) = (6,3)$$
$$\text{gcd-rule}((12,6),(12,0)) = (6,0) \qquad \text{gcd-rule}((12,3),(12,4)) = (1,0)$$

It can be seen that in the special case where all phases are zero, $\pi$ is indeed equal to the GCD of $\pi_1, \ldots, \pi_n$.

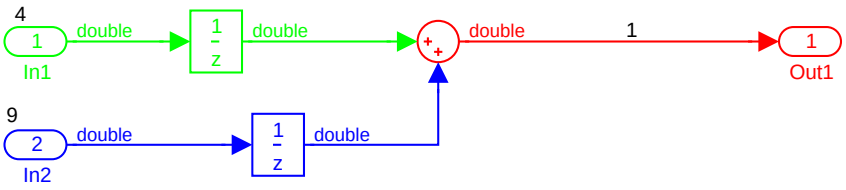



**Figure 4.4:** Illustration of the "GCD rule".

**Triggered subsystems** Another timing mechanism of SIMULINK is by means of "*triggers*". Only subsystems (not basic blocks) can be triggered. A subsystem $A$ can be triggered by a signal $x$ (of any type) in three ways, namely, "*rising*", "*falling*" or "*either*". An example of a triggered subsystem is shown in Figure 4.5. The trigger is of type "rising". Each of the three types specifies the time the trigger occurs depending on the direction with which signal $x$ "crosses" zero. For instance, the "rising" trigger occurs when $x$ changes from negative to non-negative or from non-positive to positive. However, as mentioned in the SIMULINK manual:

> "*In the case of discrete systems, a signal's rising or falling from zero is considered a trigger event only if the signal has remained at zero for more than one time step preceding the rise or fall. This eliminates false triggers caused by control signal sampling.*"

For example, as shown in Figure 4.6, a rising trigger occurs at time 6. But it does not occur at time 3 because a trigger has occurred at time 2, that is, only one time step earlier.

The sample time of blocks inside a triggered subsystem cannot be set by the user: it is inherited from the sample time $T$ of the triggering signal. The sample times of all input signals must be equal to $T$. The sample time of all outputs is $T$ as well. Thus, in the example shown in Figure 4.5, the sample times of $s, x_1, x_2$ and $y$ are all equal.

In the case of a triggered subsystem $B$ defined inside another triggered subsystem $A$, the same rules apply. First, all sample times in $B$ are inherited (cannot be set by the user). Second, the "sample times" of the trigger, inputs and outputs of $B$ must all be equal. We use the term sample time in quotes, here, because these signals have inherited sample times. In fact, the signals are updated every time $A$ is triggered, which may well be non-periodically. Still, these signals have associated timing information, namely, the triggering condition. We elaborate more on this in Section 4.5.3.

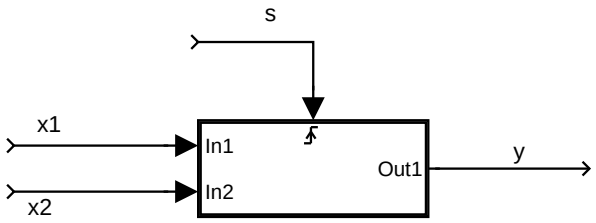



**Figure 4.5:** A triggered subsystem.

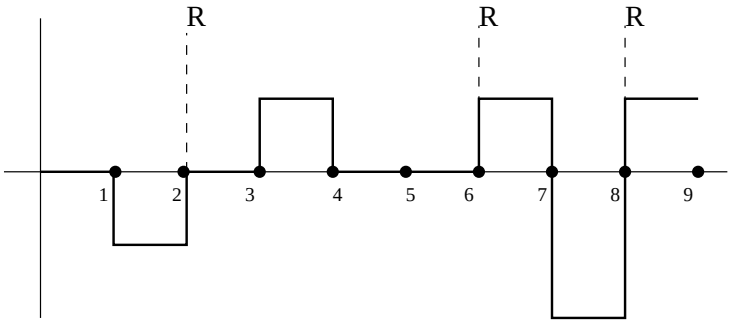



**Figure 4.6:** Illustration of the rising trigger.

**Enabled subsystems** "*Enabled*" subsystems look similar to triggered subsystems (see Figure 4.7). At first glance, the semantics of an enabled subsystem appear similar to those of a triggered subsystem as well. One would expect that the difference between the two is that the former is enabled every time the enabling signal is non-zero whereas the latter is enabled every time the triggering signal rises (or falls, or both). However, SIMULINK does not impose on enabled subsystems the restrictions on sample times that it imposes on triggered subsystems. In particular, the sample times of blocks inside an enabled subsystem do not have to be inherited. Also, the sample times of its input signals may differ from each other and may also be different from the sample time of the enabling signal. Absence of such restrictions results in quite complicated semantics for this construct.

For example, consider the model shown in Figure 4.8. It is made up of a subsystem enabled by periodic signal `e` produced by a `Pulse Generator` block – Figure 4.8(a). The enabled subsystem is simply a counter – Figure 4.8(b). The period of the counter is set to 2. This is done by setting the sample time of the $\frac{1}{z}$ block (`Unit delay` block) to 2.

Figure 4.9 shows two experiments performed with this model. In both experiments the enabling signal `e` is as shown in part (a) of the figure: it is a pulse remaining high for 3 time units and low for 3 time units, starting from high. However, in the first experiment, the sample time of `e` is set to 1, whereas in the second experiment, the sample time of `e` is set to 3.[5] As can be

[5] This is done by setting the parameters of the `Pulse Generator` block. The latter permits to obtain the

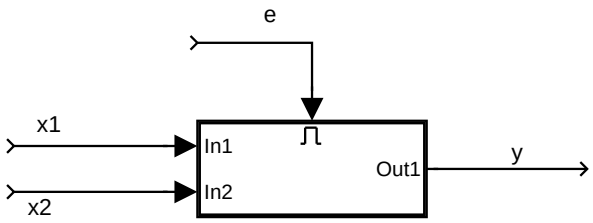
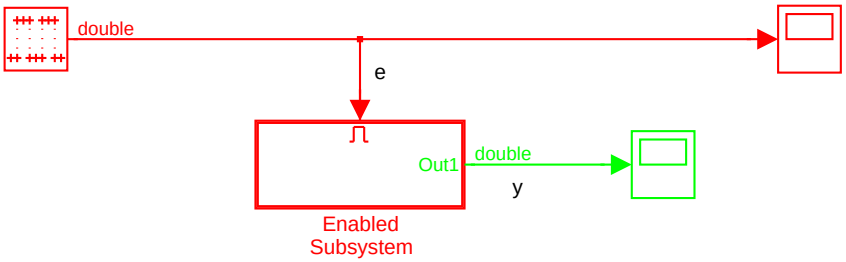
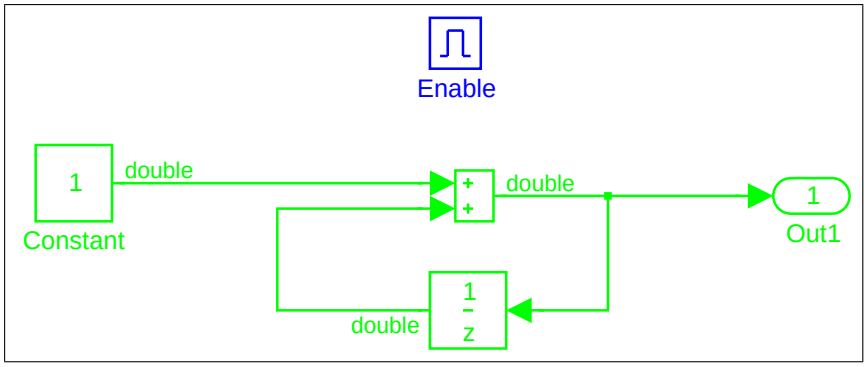



**Figure 4.7:** An enabled subsystem.




(a) A system containing an enabled subsystem.




(b) The interior of the enabled subsystem shown in (a).

**Figure 4.8:** A SIMULINK model with a counter inside an enabled subsystem.

seen, the output `y` is not the same in the two experiments. First, the initial value is not the same. Second, the signal is not updated at the same times. In particular, at time 6, it is updated when `e` has sample time 1 but it is not updated when `e` has sample time 3. This seems strange, because: (1) the pulse `e` is the same in both experiments, in particular, it is positive at time 6; (2) the "clock" of the pulse `e` "ticks" at time 6 in both experiments, since 6 is a multiple of 1 and also a multiple of 3; (3) the "clock" of the counter subsystem "ticks" at time 6 in both experiments, since 6 is a multiple of 2.

As a result of such experiments, we have not been able to identify the logic SIMULINK uses to update enabled subsystems with complex parameters. Therefore, we restrict ourselves to the subset of SIMULINK where enabled subsystems satisfy the same rules as triggered subsystems, namely: (1) all sample times of blocks inside an enabled subsystem must be set to $-1$ and (2) the sample times of all input signals must be the same and must also be equal to the sample time of the enabling signal.[6] Under these restrictions, the semantics of a subsystem $A$ enabled by signal $s$ are as expected: $A$ is activated at the beginning of each period of its inputs, provided $s$ is non-zero at that point.

**Timing modularity** Timing in SIMULINK is not as modular as in LUSTRE, in the following sense. LUSTRE nodes do not have their own time: they are only activated at instants where their inputs are active. Consequently, a node $B$ called from a node $A$ cannot be active at instants when $A$ is not active. This is true for triggered subsystems in SIMULINK as well. However, SIMULINK allows a block inside a (non-triggered) subsystem to have any sample time, possibly smaller than the parent subsystem. Thus, the block can be active while its parent system is inactive. We consider this a non-modular feature of SIMULINK. Still, we are able to translate such models in LUSTRE, by calling the parent node with the "faster" clock (on which the child block will run) and passing as parameter the "slower" clock as well.

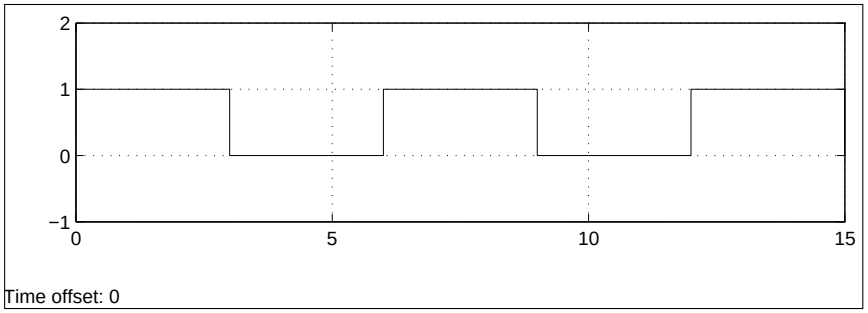
**Timing rules** Another difference between SIMULINK and LUSTRE lies in how changes of timing are performed in the two languages. In LUSTRE, this can only be done using the `when` and `current` operators. In SIMULINK, the sample time of a signal can be changed using the `Unit Delay` block or the `Zero-order Hold` block. In order to do this, however, the following rules must be obeyed:[7]

1. *"When transitioning from a slow to fast rate, a Unit Delay running at the slow rate must be inserted between the two blocks. When transitioning from a fast to a slow rate, a Zero Order Hold running at the slow rate must be inserted between the two blocks."*

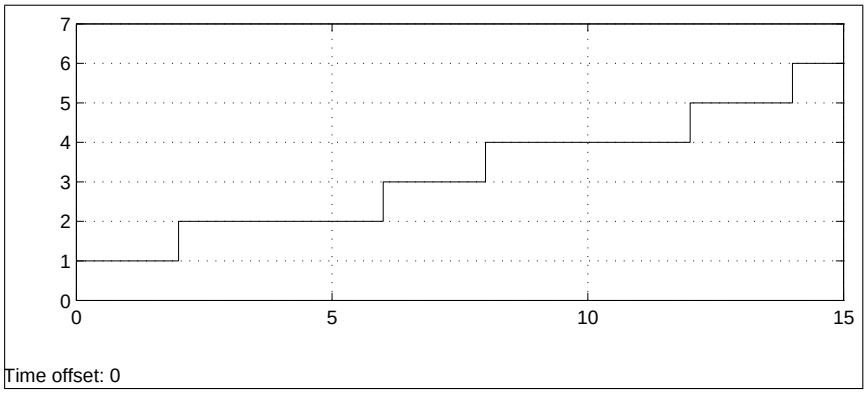
same signal using different parameter settings, in particular, setting the "sample time", "period", etc., parameters.

[6] SIMULINK itself issues warnings in various cases where condition (2) is not satisfied, for instance: "*Warning: Sample time of enable signal feeding subsystem block 'enable_problem/Enabled Subsystem' is slower than blocks within the enabled subsystem. This can result in nondeterministic behavior in a multitasking real-time system. Consider adding a Rate Transition block followed by a Signal Specification block with a sample time equal to the enable signal rate.*"

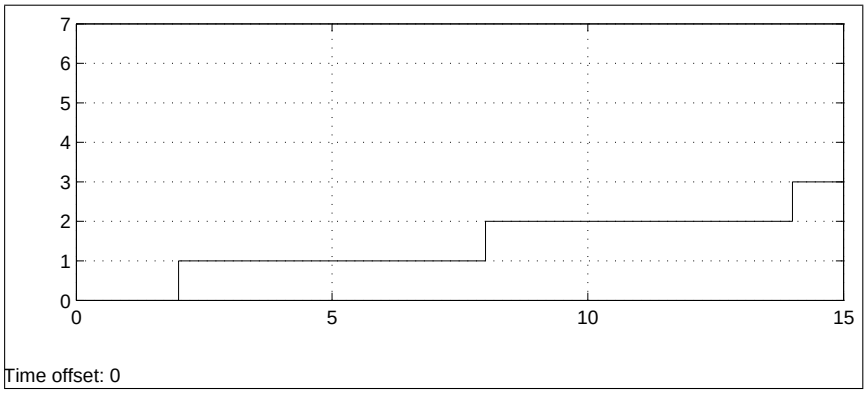
[7] The rules are quoted from error messages produced by the SIMULINK tool.




(a) The enabling signal ∈.




(b) The output y when enable ∈ has period 1.




(c) The output y when enable ∈ has period 3.

**Figure 4.9:** A strange behavior of the model of Figure 4.8.

2. *"Illegal rate transition found involving Unit Delay block ... When using this block to transition rates, the input must be connected to the slow sample time and the output must be connected to the fast sample time. The Unit Delay must have a sample time equal to the slow sample time and the slow sample time must be a multiple of the fast sample time. Also, the sample times of all destination blocks must be the same value."*

3. *"Illegal rate transition found involving Zero Order Hold block ... When using this block to transition rates, the input must be connected to the fast sample time and the output must be connected to the slow sample time. The Zero Order Hold must have a sample time equal to slow sample time and the slow sample time must be a multiple of the fast sample time. Also, the sample times of all source blocks must be the same value."*[8]

The necessity of these rules is dictated by implementation concerns. Indeed, for simulation purposes the rules are not necessary, since time is logical and communication between tasks can be delayed arbitrarily so that the reader task reads the correct data. However, using the REAL-TIME WORKSHOP code generator for a real-time execution platform raises concerns about the data integrity in case of preemption.

For example, if a slow-rate block $S$ "feeds" its output to a fast-rate block $F$, and the worst-case execution time (WCET) of $S$ is greater than the period of $F$, then $S$ cannot communicate its output directly to $F$ ($S$ would have to complete before $F$ can start, but this exceeds the period of $F$). A solution is to insert a unit-delay between $S$ and $F$. Then $F$ only needs the *previous* value of $S$ and does not have to wait for the *current* instance of $S$ to complete. The latter can execute concurrently with current instances of $F$ using a multi-tasking implementation scheme on top of an operating system providing a preemptive scheduler.

Unfortunately, such rules create confusion among specification and implementation concerns. There is *a priori* no reason why the class of admitted specifications should be restricted because of implementation concerns. Such concerns may not even be an issue: for instance, when the WCETs of $S$ and $F$ "fit" into the period of $F$. Also, implementation constraints depend on decisions such as the choice of hardware which are likely to change over time.

We believe that in a model-based approach specification and implementation should be clearly separated. LUSTRE does separate the two. First, it imposes no unnecessary restrictions on the specification side. In the case of the above example, it offers the designer the possibility to model both cases, with or without a unit-delay between $S$ and $F$. Naturally, the choice of the designer may be influenced by implementation concerns. Second, different implementation techniques are available. As mentioned in the introduction, these include the traditional single-processor, single-tasking code generation methods, plus more recent methods such as those for the Time Triggered Architecture [CCM+03] or for a single-processor, multi-tasking architecture that we present in Chapter 7.

After this short digression, let us return to the timing rules of SIMULINK. The second rule above explains why the model of Figure 4.3 is rejected when the `Gain` block is present and accepted otherwise. Indeed, after computing the sample times according to the method described

[8] The third rule may seem strange since it implies that a zero-order hold block can have more than one inputs. In fact, this happens when the input to this block comes from a `Mux` block, thus, encoding a vector of signals.

in the following section, we find that the output of block `Unit Delay2` has sample time 3. This output is "fed" to the `Gain` block, which also has sample time 3, and to the left-most `Sum` block, which has sample time 1 (the greatest common divisor of 2 and 3). This violates the last part of the second rule, namely, that "the sample times of all destination blocks must be the same".

As mentioned earlier, SIMULINK requires that for all sample times $(\pi, \theta)$ in the model, $\pi$ is not smaller than $\theta$. This rule may be violated during gcd-rule operations, in which case SIMULINK issues an error.

**Summary** In summary, the timing mechanism in SIMULINK can be described as follows. SIMULINK has a set of sample times,

$$SampleTimes = \{-1\} \cup \{(\pi, \theta) \mid \pi, \theta \in \mathsf{Q}_{\geq 0}, \pi \geq \theta, \pi \neq 0\},$$

where $\mathsf{Q}_{\geq 0}$ is the set of non-negative rationals. Similarly to a type signature, a SIMULINK block can be given a sample time signature, as shown in Table 4.4. The terms "Output", "Math" and so on refer to the blocks of the corresponding libraries shown in Figure 4.1. The notation "Discrete$_\beta$" refers to a block of the Discrete library which has its sample time set to $\beta$ by the user (note that default is $-1$). The notation "Triggered$_\alpha$" (respectively, "Enabled$_\alpha$") refers to a triggered (respectively, enabled) subsystem where the triggering (respectively, enabling) signal has sample time $\alpha$. We can see that:

- An Output block preserves the sample time of its input.[9]
- All blocks in the "Math" library produce an output having sample time computed by the GCD rule. The same is true for the Switch block.
- For a block of the "Discrete" library, there are two cases. If the block has its sample time parameter set to $-1$, the output has the same sample time as the input, otherwise, it has the sample time set in the block.
- Triggered (respectively, enabled) subsystems require that all inputs have the same sample time, which must also be equal to the sample time of the triggering (respectively, enabling) signal. This sample time is preserved on the outputs.

### 4.5.3 Clock Inference

Similarly to type inference, the objective of clock inference is to compute, for each signal in the SIMULINK model, the timing information associated with this signal. This timing information is then used during the translation step to generate the clock of the corresponding variable in the generated LUSTRE program. An additional objective of clock inference is to define the period (and initial phase) at which the LUSTRE program must be run, in order to respect the real-time semantics of the original SIMULINK model.

[9] Strictly speaking, an Output block has no output (the output is provided to the parent system of the subsystem where the block is defined). However, it is useful to view this block as the identity function.

$$
\begin{array}{rcl}
\text{Output} & : & \alpha \rightarrow \alpha, \alpha \in SampleTimes \\
\text{Math} & : & \alpha_1 \times \cdots \times \alpha_n \rightarrow \mathsf{gcd\text{-}rule}(\alpha_1, ..., \alpha_n), \alpha_i \in SampleTimes, i = 1, ..., n \\
\text{Switch} & : & \alpha \times \beta \times \gamma \rightarrow \mathsf{gcd\text{-}rule}(\alpha, \beta, \gamma), \alpha, \beta, \gamma \in SampleTimes \\
\text{Input} & : & \alpha \rightarrow \alpha, \alpha \in SampleTimes \\
\text{Discrete}_\beta & : & \alpha \rightarrow \alpha, \alpha \in SampleTimes, \beta = -1 \\
\text{Discrete}_\beta & : & \alpha \rightarrow \beta, \alpha \in SampleTimes, \beta \in SampleTimes, \beta \neq -1 \\
\text{Triggered}_\alpha & : & \alpha \times \cdots \times \alpha \rightarrow \alpha, \alpha \in SampleTimes \\
\text{Enabled}_\alpha & : & \alpha \times \cdots \times \alpha \rightarrow \alpha, \alpha \in SampleTimes
\end{array}
$$

**Table 4.4:** Sample time signatures in SIMULINK.

**Clock inference of triggered and enabled subsystems** Consider a SIMULINK signal $x$. If $x$ is defined inside a triggered or enabled subsystem $A$ then the timing information we want to associate with $x$ is, much like in LUSTRE, the Boolean signal $b$ which represents the times $A$ is activated. Signal $b$ is not explicitly defined in the SIMULINK model. It is implicitly defined by the signal $s$ feeding the trigger or enabled icon in the model, the type of $s$ and the type of the trigger. Thus, no particular inference must be performed in this case. How to explicitly construct $b$ in the generated LUSTRE program is explained in Section 4.6.

**Clock inference of sample times** Now, consider the case where $x$ is defined neither inside a triggered subsystem nor inside an enabled subsystem. In this case, the timing information of $x$ is a sample time. Although the latter can be undefined (i.e., the default sample time $-1$) we might still have some information which is useful to keep. For instance, when $x$ is the output of a `Sum` block, we know that the sample time of $x$ will be related to the sample times of the inputs of the block according to the GCD rule. We would like to infer this information automatically from the SIMULINK model.

We do this using what can be qualified a "symbolic" technique, as follows. We consider the language of *clock types* defined by the following syntax:

$$t ::= \alpha \mid (\pi, \theta) \mid \mathsf{gcd}(t, t)$$

where $t$ is a clock-type term, $\alpha$ is a clock-type variable in a given set of variables, $(\pi, \theta) \in SampleTimes \setminus \{-1\}$ is a clock-type constant and $\mathsf{gcd}(t, t)$ is a composite term corresponding to the gcd-rule operation. Using arithmetic properties of the latter such as associativity and the properties of GCD, clock-type terms can be sometimes simplified. For instance, we can write $\mathsf{gcd}(t_1, t_2, t_3)$ instead of $\mathsf{gcd}(t_1, \mathsf{gcd}(t_2, t_3))$ and $(1, 0)$ instead of $\mathsf{gcd}((2, 1), (3, 0))$. Moreover, if we fix an order (say, lexicographic) on the set of clock-type variables then we can define for each clock-type term its *canonical form*, obtained by (1) eliminating all but one constants by applying gcd-rule, (2) eliminating multiple occurrences of the same subterm (since

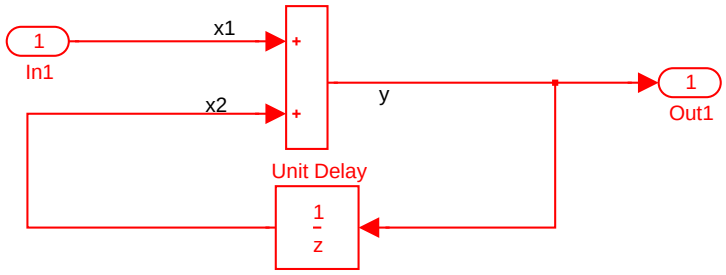



**Figure 4.10:** Example of clock inference.

$\mathsf{gcd}(t, t, t') = \mathsf{gcd}(t, t')$), (3) ordering the variables according to the given order and (4) ordering constants before variables. For example, if variables are ordered $\alpha_1, \alpha_2, ...$, then the canonical form of $\mathsf{gcd}(\alpha_2, \alpha_1, (2, 0), (3, 0))$ is $\mathsf{gcd}((1, 0), \alpha_1, \alpha_2)$.

Given a SIMULINK model we infer for each signal in the model a clock-type which is a term in the syntax above. We treat triggered and enabled subsystems as "black boxes", that is, we do not infer clock-types for the signals defined inside these subsystems. This is justified because all such signals will have inherited sample times and we can associate to them LUSTRE clocks using the method described above. For the rest of the signals, we proceed as follows.

First, we assign a clock-type variable to each signal. Then, we build a set of equations relating these variables, according to the signatures given in Table 4.4. For example, if $y$ is the output of a `Sum` block with inputs $x_1, x_2$ and $\alpha_1, \alpha_2, \alpha_y$ are the corresponding clock-type variables, we get the equation

$$\alpha_y = \mathsf{gcd}(\alpha_1, \alpha_2).$$

We then solve this set of equations using a unification algorithm, as in the case of type inference. However, it is worth noting two particularities of clock-type unification. First, when two terms $f(t)$ and $g(t')$ must be unified but $f \neq g$, unification fails. This cannot occur in the case of clock types since we have a single operator $f$, namely, $\mathsf{gcd}$. Second, when variable $\alpha$ must be unified with $f(t)$ but $\alpha$ occurs free in $t$, standard unification fails. However, clock-type unification will succeed, because of special properties of the $\mathsf{gcd}$ operator. In general, $\alpha = \mathsf{gcd}(t)$ has a solution $\alpha = \mathsf{gcd}(t_{-\alpha})$ where $t_{-\alpha}$ is obtained by eliminating $\alpha$ from $t$.

To see this, consider the example shown in Figure 4.10. We assume that the `Unit Delay` block has sample time set to default, that is, $-1$. From this model we get the equations

$$\alpha_y = \mathsf{gcd}(\alpha_1, \alpha_2), \qquad \alpha_2 = \alpha_y$$

thus also

$$\alpha_2 = \mathsf{gcd}(\alpha_1, \alpha_2).$$

As said above, the solution is to eliminate $\alpha_2$ from the right-most term, obtaining

$$\alpha_2 = \mathsf{gcd}(\alpha_1)$$

or

$$\alpha_2 = \alpha_1.$$

Thus, unification succeeds in this case with $\alpha_1 = \alpha_2 = \alpha_y$.

Clock-type unification may fail, for instance, when trying to unify constant clock-types that differ. One situation where this occurs is when two inputs of, say, a triggered subsystem have different sample times, thus, violating the signature shown in Table 4.4. When unification fails, the SIMULINK model is rejected due to timing errors.

The solution provided by the unification algorithm when the latter succeeds is to be interpreted as follows. For signals with inferred clock-types which are constants $(\pi, \theta)$, we know exactly their sample time. For signals with inferred clock-types which are variables $\alpha_1, \alpha_2, ...$, their sample times are unknown and can be anything. The rest of the signals will have a clock-type of the form $\mathsf{gcd}(\alpha_i, \alpha_j, ...)$ or $\mathsf{gcd}((\pi, \theta), \alpha_i, \alpha_j, ...)$. The sample times of such signals are not known, however, they are subject to constraints expressed by the GCD rule.

## 4.6 Translation

The type and clock inference steps are independent and can be performed in any order. Once this is done, the translation itself is performed, in a hierarchical manner. The SIMULINK model is organized as a tree, where the children of a subsystem (or system) are the subsystems (or blocks) directly appearing in it. The translation is performed following this hierarchy in a bottom-up fashion, that is, starting from the basic blocks which form the leaves of the hierarchy tree. SIMULINK signals are mapped into LUSTRE flows. SIMULINK blocks are translated as input/output equations between signals, either by using predefined LUSTRE operators, in the case of simple blocks, or by calling LUSTRE nodes, which capture the functionality of more complex blocks. SIMULINK subsystems are also translated by declaring and calling LUSTRE nodes.

**Naming** In accordance with the goal of traceability and in order to preserve the hierarchical structure of the SIMULINK model in the generated LUSTRE program, each LUSTRE node is named with the corresponding path of names in the SIMULINK tree. For example, a subsystem $B$ contained in a subsystem $A$ will be translated into a LUSTRE node named `NameofA_B` where `NameofA` is the name of the LUSTRE node corresponding to subsystem $A$.

When a signal in SIMULINK has a name, the name is preserved during the translation. However, it is often the case that signals in SIMULINK are not named (they are simply "wires" connecting two blocks). In such a case, the name given to the corresponding LUSTRE variable reflects the block which produces the signal. For example, if an unnamed signal is produced by a block named `Sum1` then the corresponding LUSTRE variable will be named `Sum1_out`.

**Translation of basic SIMULINK blocks** Simple SIMULINK blocks are translated into predefined LUSTRE operators. In particular:

- The `Sum` block is translated using the + and − LUSTRE operators.

- The `Product` and `Gain` blocks are translated using the $\star$ and / LUSTRE operators and constants.

- The `Logical Operator` block is translated using the Boolean LUSTRE operators `and`, `or`, `not`.

- The `Relational Operator` block is translated using the LUSTRE comparison operators <, >, <=, >=, =, <>.

- The `Unit Delay` block is translated using the LUSTRE operators `pre` and `->`. In particular, if `x` and `y` are the input and output of the block and `init` is the initial value (a constant) specified in the dialog box of the block, then the following LUSTRE code is generated:

```
y = init -> pre(x) ;
```

- A `Zero-Order Hold` block with sample time set to $-1$ is the identity function, thus no special code needs to be generated. If the sample time of the block is set, then the block is translated using a `when` statement, possibly preceded by a `current` statement. The latter is needed in the case where the input signal is not running on the basic clock. For example, consider the model of Figure 4.13 and assume that the sample time of the `Zero-Order Hold` block is 3. Thus, the output has sample time 3 as well. First, suppose that the sample time of the input is 1 and that this also corresponds to the basic clock. Then, the translation would be:

```
y  = x when clock_3_0 ;
```

  Second, suppose that the sample time of the input is 2, in which case the sample time corresponding to the basic clock is 1. Then, the translation would be:

```
x_ = if clock_2_0 then current(x) else 0 -> pre(x_) ;
y  = x_ when clock_3_0 ;
```

- A `Constant` block is translated into the corresponding constant. Here, the type information of the output of the block is used. For example, the output of constant block 0 is translated to `false` if it has type `boolean`, to `0` if it has type `int` and to `0.0` if it has type `real`. This is because in LUSTRE constants are not overloaded with many types.

- The `Saturation` block truncates its input according to bounds provided by the user. It is translated using `if-then-else` statements. For instance, if the upper and lower bounds are 0.5 and $-0.5$ then the generated LUSTRE code is like:

```
Saturation = if In1 > 0.5  then  0.5
             else if In1 < -0.5 then -0.5
             else In1 ;
```

- A `Switch` block is also translated using an `if-then-else` statement, where the middle input is compared with the value of the `threshold` block property to select one of the two other inputs as the output. For instance, if the threshold is set to `5.0` then the generated LUSTRE code is like:

```
Switch = if (In2 >= 5.0) then In1 else In3 ;
```

More complex SIMULINK blocks are translated into LUSTRE nodes. In particular:

- The `Pulse Generator` is translated with the use of `pre` operators and recursion. For example, the pulse with $amplitude = 10$, $period = 5$, $phase = 2$, $pulse\_width = 2$ will be translated to the following LUSTRE node:

```
node PulseGen_1()
returns (out: real);
var dpg1, dpg2, dpg3, dpg4, dpg5, pha1, pha2 :real ;
let
    dpg1 = 10.0 -> pre dpg2 ;
    dpg2 = 10.0 -> pre dpg3 ;
    dpg3 =  0.0 -> pre dpg4 ;
    dpg4 =  0.0 -> pre dpg5 ;
    dpg5 =  0.0 -> pre dpg1 ;

    pha1 =  0.0 -> pre dpg1 ;
    pha2 =  0.0 -> pre pha1 ;
    out  =  pha2 ;
tel
```

  The five first equations define a signal which repeats a cycle of five instants, having value `10.0` the first two times and zero the rest three times. The next two equations "shift" the signal to the correct phase.

- The `Discrete Filter` and `Discrete Transfer Function` blocks are translated into nodes using arithmetic operators and `pre` according to standard algebraic manipulations of the expression specified in the dialog box of the block. For example, the `Discrete Transfer Function` block with expression $\frac{z+2}{z^2+3z+1}$ is translated into the LUSTRE node:

```
node Transfer_Function_3(x: real) returns(y: real);
var y_1, y_2: real;
let
    y   = 0.0 -> pre(y_1) ;
    y_1 = x - 3.0*y + (0.0 -> pre(y_2)) ;
    y_2 = 2.0*x - y ;
tel.
```

  This comes from the interpretation $y = \frac{z+2}{z^2+3z+1}x$, that is, $(z^2+3z+1)y = (z+2)x$, which yields $y = z^{-1}x + 2z^{-2}x - 3z^{-1}y - z^{-2}y = (x-3y)z^{-1} + (2x-y)z^{-2}$.

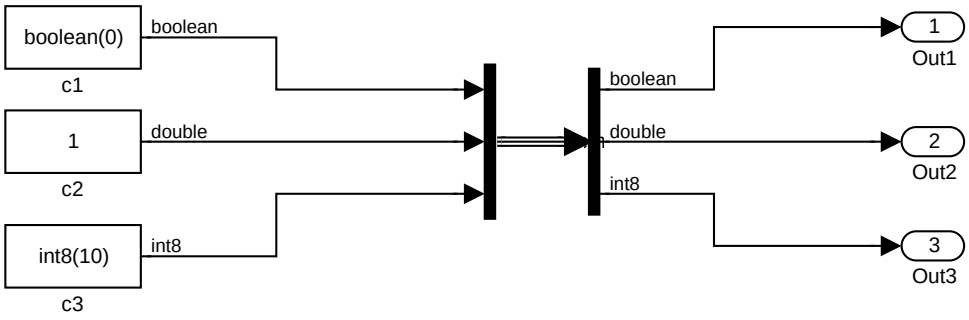



**Figure 4.11:** Mux - Demux example.

- The `Data Type Conversion` block is translated as a *type casting* operation. For example, the following code can be used to turn a real `x` into a boolean `b`:

```
node real_to_bool(x: real)
returns (b: bool) ;
let
    b = if (x = 0.0) then true else false;
tel
```

- The `Mux` and `Demux` blocks permit to group and ungroup signals into "bundles", for example, as shown in Figure 4.11. Signals of different types are allowed. Thus, a group of signals can be viewed as a *record*. The translation to LUSTRE starts by declaring a new type composite type from the types of each signal in the group (known after type inference). Then, LUSTRE `FROM` and `TO` are used to compose/decompose signals. For the example of Figure 4.11, the LUSTRE code will be as shown below:

```
type type_bri = {bool, real, int} ;

node mux(c1: bool; c2: real; c3: int)
returns(out: type_bri) ;
let
    out = TO(type_bri; c1, c2, c3) ;
tel


node demux(in: type_bri)
returns (Out1: bool; Out2: real; Out3: int);
let
    (Out1, Out2, Out3) = FROM (type_bri; in) ;
tel
```

- The `Combinatorial Logic` block implements a truth table. It can be implemented using `if-then-else` statements.

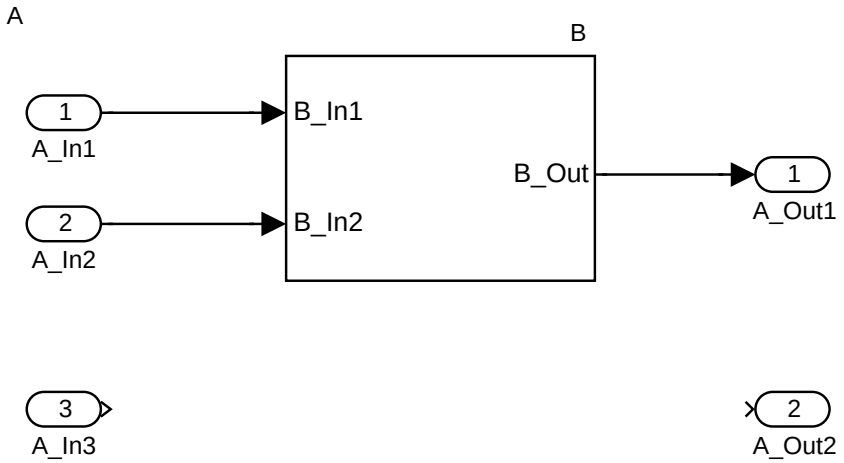



**Figure 4.12:** SIMULINK system $A$ with subsystem $B$.

**Translation of subsystems and subsystem calls** A SIMULINK subsystem is translated into a LUSTRE node, possibly containing calls to other nodes. The LUSTRE node has the same inputs and outputs as the SIMULINK subsystem, plus, sometimes, the clock of some of its inputs (this is done for modularity). Here is an example of such a translation. Consider the SIMULINK model shown in Figure 4.12, with subsystems $A$ and $B$. The LUSTRE code generated for this example is as shown below:

```
node A(A_in1, A_in2, A_in3 : real)
returns (A_out1, A_out2 : real) ;
let
      A_out1 = B(A_in1, A_in2) ;
      A_out2 = ...
tel


node B(B_in1, B_in2 : real)
returns (B_out : real) ;
...
```

An example where it is necessary to pass clock information as input to the LUSTRE node is given in Section 6.2.1.

**Translation of triggered subsystem calls** If the subsystem to be called is a triggered subsystem such as the one shown in Figure 4.5, special code needs to be added in order to capture the timing behavior according to the trigger. We illustrate how this is done in the case of a "rising" trigger. The translation method is similar for the other two types of triggers.

First, as mentioned in Section 4.5.3, we must explicitly define in LUSTRE the Boolean flow representing the trigger, which is implicit in SIMULINK. This is performed using three auxiliary

LUSTRE nodes `rising_bool`, `rising_int` and `rising_real`. These nodes are independent from the SIMULINK model and can be predefined in a library. There is one node for each possible type of the triggering signal $s$, because LUSTRE does not currently support polymorphism. We only present `rising_int` here. The other two nodes are defined similarly.

```
node rising_int (s : int)
returns (rise_trigger : bool);
var neg_to_nonneg, nonpos_to_pos, not_before : bool ;
let
    rise_trigger  = neg_to_nonneg or (nonpos_to_pos and not_before) ;
    neg_to_nonneg = false -> (pre(s < 0) and (s >= 0)) ;
    nonpos_to_pos = false -> (pre(s <= 0) and (s > 0)) ;
    not_before    = false -> not pre(rise_trigger) ;
tel
```

The node takes as input an integer flow and returns a Boolean flow which is true whenever there is a rising trigger on $s$. The local variables `neg_to_nonneg` and `nonpos_to_pos` represent the situations where the signal `s` rises from negative to zero or positive and from negative or zero to positive, respectively. The local variable `not_before` ensures that in situations such as the one shown in Figure 4.6 there is no trigger produced when it should not.

Then, the LUSTRE code generated when calling a triggered subsystem `A` like the one of Figure 4.5 is as follows (we assume that variables `x1` and `x2` are of type `int`):

```
var trig : bool;

trig = rising_int(s) ;
x1t  = x1 when trig ;
x2t  = x2 when trig ;
yt   = A(x1t, x2t) ;
y    = if trig then current(yt) else (0 -> pre(y)) ;
```

Variables `x1t` and `x2t` are obtained by "sampling" the inputs only at times dictated by the trigger. Consequently, node `A` is activated only at those times as well, and its output `yt` has the same clock, that is, `trig`. To obtain output `y` which must have the same clock (sample time) of the inputs `x1` and `x2`, we must perform a `current` operation. However, a simple `current` is not enough, since it may leave `y` undefined for the initial instants when the trigger is potentially false. This is why we use the `if-then-else` construct in the equation defining `y`. This construct ensures that the value of `y` is equal to 0 until the first time `trig` becomes true.

**Translation of enabled subsystem calls** The LUSTRE code generated when calling an enabled subsystem is the same as when calling a triggered subsystem, except that flow `trig` is replaced by flow `enab`, defined as follows:

```
var enab : bool;

enab = if (e > 0) then true else false ;
```

where `e` is the signal feeding the `Enable` icon.

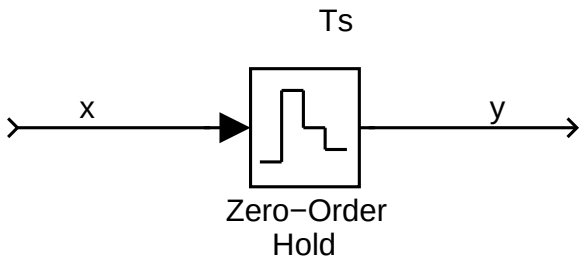



**Figure 4.13:** A `Zero-Order Hold` block modifying the period of its input.

**Translation of signals with sample times** The unification algorithm on sample times, presented in Section 4.5.3, produces a "symbolic" sample time for each SIMULINK signal. It remains to associate these symbolic sample times to LUSTRE clocks. Here, a problem arises, due to the unknown sample times. The problem comes from the fact that a LUSTRE program has a single basic clock and all other clocks are subclocks of the basic clock. Thus, in LUSTRE, $\alpha_1$ and $\alpha_2$ must be subclocks of gcd$(\alpha_1, \alpha_2)$, since the latter is "faster". To solve this problem, we take a pragmatic approach. We assume one of the following (this becomes an option to the translation algorithm):

- either that all unknown sample times are equal to the gcd-rule of all known sample times,
- or that all unknown sample times are fixed by the user when generating the LUSTRE code.

In both cases the sample times become known at the moment of code generation. The basic clock of the LUSTRE program is assumed to have period and initial phase equal to the gcd-rule of all sample times in the system. For each other sample time, a boolean flow corresponding to this sample time is constructed inside the LUSTRE program and used to create flows associated with this sample time. For example, if $(2, 1)$ and $(3, 0)$ are the only sample times in the model, then the basic clock will be assumed to have sample time $(1, 0) =$ gcd-rule$((2, 1), (3, 0))$. In the main node of the LUSTRE program, the following clocks will be created:

```
var clock_2_1: bool, clock_3_0: bool;
var cnt3: int;

clock_2_1 = false -> not pre(clock_2_1) ;

cnt3      = 0 -> (pre(cnt3) + 1) mod 3;
clock_3_0 = if (cnt3 = 0) then true else false ;
```

Clock `clock_2_1` alternates between false and true, starting at false, to model the initial phase 1 and the period 2. Clock `clock_3_0` follows the cycle true, false, false, modeling a period 3 with initial phase zero. This clock is defined using a counter modulo three.

In general, a clock with period `per` and phase `ph` is produced using the following node:

```
node make_clock(per: int; ph: int)
returns( clock: bool );
var cnt: int;
let
    cnt   = (per - ph) -> (pre(cnt) + 1) mod period ;
    clock = if (cnt = 0) then true else false ;
tel
```

The above clocks are used during LUSTRE code generation as follows. Consider the Zero-Order Hold block of Figure 4.13 and assume that the sample time of input $x$ is $(1, 0)$ and the sample time set to the zero-order hold block is $(2, 1)$. Then the sample time of the output $y$ is also $(2, 1)$ and the generated LUSTRE code is as follows:

```
y = x when clock_2_1 ;
```

If, instead of a Zero-Order Hold block we had, say, a Discrete Transfer Function block, the LUSTRE code would be:

```
y = DTF(x when clock_2_1) ;
```

where `DTF` is the LUSTRE node implementing the Discrete Transfer Function block.

Now consider the `Sum` block of Figure 4.4 and assume that the sample times of inputs $x_1$ and $x_2$ are $(2, 1)$ and $(3, 0)$, respectively. Also assume as a first case that the basic clock is $(1, 0)$. Then the sample time of the output $y$ is $(1, 0)$ and the generated LUSTRE code is as follows:

```
x1t = if clock_2_1 then current(x1) else (0 -> pre(x1t)) ;
x2t = if clock_3_0 then current(x2) else (0 -> pre(x2t)) ;
y   = x1t + x2t ;
```

`x1` and `x2` have been previously produced using the `when` operator on the appropriate clocks. Then, `x1t` and `x2t` are on the basic clock and so is the output `y`. The `if-then-else` construct is used with the `current` operator as previously, to ensure well-defined initial values.

As a second case, assume that $(1, 0)$ is not the basic clock but a subclock. Then `y` must be obtained as

```
y   = (x1t + x2t) when clock_1_0 ;
```

## 4.7 Related Work

Before concluding this Chapter, we comment on some work related to its contents.

[BFM+05] study the translation between SIMULINK and the language supported by the ASCET tool-set by the ETAS group. ASCET's language is closer to the implementation level than LUSTRE, in the sense that the basic entities are tasks scheduled by a real-time operating system (RTOS).

[TNTBS00] report on an approach to co-simulate discrete controllers modeled in the synchronous language Signal [GGBM91] along with continuous plants modeled in SIMULINK. [CHLrA02] present tools for co-simulation of process dynamics, control task execution and network communication in a distributed real-time control system. [SBCR01] use a model-checker

to verify a SIMULINK/STATEFLOW model from the automotive domain, however, they translate their model manually to the input language of the model-checker.

A number of approaches are based in extending SIMULINK with libraries of predefined blocks and then using SIMULINK as a front-end or simulator. The hybrid-system model-checker CheckMate uses such an approach [SRKC00, CK03]. [KSHP02] extend SIMULINK with the capability of expressing designs in the time-triggered language Giotto [HHK01].

[JZW+00] report on translating SIMULINK to the *SPI model*, a model of concurrent processes communicating with FIFO queues or registers. The focus seems to be the preservation of value over-writing which can occur in multi-rate systems when a "slower" node receives input from a "fast" one.

[JB03] report on MAGICA, a type-inference engine for Matlab. The focus is on deriving information such as whether variables have real or imaginary values, array sizes for non-scalars, and so on.

[ASK04] propose a method to translate SIMULINK/STATEFLOW models into hybrid automata using graph transformations.

## 4.8 Conclusions

We have presented a method for translating a discrete-time subset of SIMULINK models into LUSTRE programs. The translation is done in three steps: type and clock inference, followed by a hierarchical bottom-up translation. We have implemented the method in a tool called S2L and applied it to two embedded controller applications from the automotive domain. The interest of our tool is that it opens the way to the use of formal and certified verification and implementation tools attached to the LUSTRE tool chain. Also, in the process of translation, we explained and formalized the typing and timing mechanisms of SIMULINK.

Perhaps the most significant drawback of our approach is its dependency on syntax and semantics of SIMULINK models. New versions of SIMULINK appear as often as every six months and sometimes major changes are made with respect to previous versions. This situation seems difficult to avoid given the relative "monopoly" of SIMULINK/STATEFLOW in the control design landscape. Another weakness of our tool is its incompleteness: several unsafe constructs of SIMULINK are not translated. Yet this can be seen as the price to pay for having a sound translation.

# Chapter 5

# Analysis and Translation of STATEFLOW to LUSTRE

In this Chapter we will analyze STATEFLOW and go through a faithful translation towards LUSTRE, trying to preserve the semantics that we define in the next Sections.

Before we can attempt to define which features of STATEFLOW are suitable for translation into LUSTRE, we have to illustrate some of the semantical issues with STATEFLOW, which are also likely to cause problems with our translator. These issues range from "serious" ones, such as non-termination of a simulation step or stack overflow, to more "minor" ones, such as dependence of the semantics upon the positions of objects in the STATEFLOW diagram. First, we briefly describe the STATEFLOW language and informally explain its semantics (for a formal semantics, see [HR04]).

## 5.1 A short description of STATEFLOW

STATEFLOW is a graphical language resembling Statecharts [Har87]. The semantics of STATEFLOW are embodied in the interpretation algorithm of the STATEFLOW simulator[1]. A STATEFLOW *chart* has a hierarchical structure, where states can be refined into either *exclusive (OR)* states connected with transitions or *parallel (AND)* states, which are not connected. It is important to note that parallel states are not executed concurrently, but sequentially. Figure 5.13 shows an example: $A$ and $B$ are parallel states (with *parent* the root state), while all their *child* states are exclusive. A transition can be a complex (possibly cyclic) flow graph made of *segments* joining connective *junctions*. Each segment can bear a complex label with the following syntax (all fields are optional):

$$E[C]\{A_c\}/A_t$$

where $E$ is an *event*, $C$ is the *condition* (i.e., guard), $A_c$ is the *condition action* and $A_t$ is the *transition action*. $A_c$ and $A_t$ are written in the *action language* of STATEFLOW, which contains

[1] This is documented in a 900-page long User's Guide STATEFLOW and STATEFLOW Coder, User's Guide, Version 5. Available at http://www.mathworks.com/products/stateflow/ From this guide we have borrowed our terminology.

assignments, emissions of events, and so on. The order of execution of those actions is stated later. Actions written in the action language can also be associated to states. A state can have an *entry action*, a *during action*, an *exit action* and *on event $E$ actions*, where $E$ is an event.

The interpretation algorithm is triggered every time an event arrives from SIMULINK or from within the STATEFLOW model itself.[2] The algorithm then executes the following steps:

**Search for active states:** this search is performed hierarchically, from top to bottom. At each level of hierarchy, when there are parallel states, the search order is a *graphical two dimensional one: states are searched from top to bottom and from left to right*, in order to impose determinism upon the STATEFLOW semantics.

**Search for valid transitions:** once an active state is found, its output transitions are searched for a valid one to follow, with respect to several criteria: the event of the transition must be present and its condition must be true. As mentioned earlier, both the event on a transition and the condition are optional, in which case those criteria are not checked. The goal is to find a transition which is *valid* all the way from the source state to the destination state. In particular, when the transition is multi-segment, the condition actions of each segment are executed while searching and traversing the transition graph, even if the condition does not hold. The search order is again deterministic: transitions are searched according to the *12 o'clock rule*.[3]

**Execute a valid transition:** once a valid transition is found, STATEFLOW follows these steps: execute the exit action of the source state, set the source state to inactive, execute the transition actions of the transition path, set the destination state to active and finally execute the entry action of the destination state.

**Idling:** when an active state has no valid output transitions an active state performs its during action and the state remains active.

**Termination:** occurs when there are no active states.

It should be emphasized that each of the executions *runs to completion* and this makes the behavior of the overall algorithm very complex. In particular, *when any of the actions consists of broadcasting an event, the interpretation algorithm for that event is also run to completion before execution proceeds.* This means that the interpretation algorithm is recursive and uses a *stack*. However, as we will see, the stack does not store the full state, which leads to problems of side effect (Section 5.2). Also, without care, the stack may overflow (Section 5.2).

### Interface between SIMULINK model and STATEFLOW chart

Every STATEFLOW model resides inside a SIMULINK system and more precisely it is introduced, in the SIMULINK model, using the "Chart" block from the "sflib" library (that may be invoked

[2] The SIMULINK event is often a SIMULINK*trigger*, although it can also be the simulation step of the global SIMULINK-STATEFLOW model.

[3] Notice that this is considered harmful even in the STATEFLOW documentation, where it is stated: "Do not design your STATEFLOW diagram based on the expected execution order of transitions."

with the `stateflow` command in the MATLAB command line environment). STATEFLOW charts run as blocks in a SIMULINK model. The STATEFLOW block connects to other blocks in the model by input and output signals, the same signals that are used for the interconnection between all the other SIMULINK blocks. Through these connections, STATEFLOW and SIMULINK share data and respond to events that are broadcasted between model and Chart.

In fact, STATEFLOW is updated (the interpretation algorithm is executed), as we saw earlier, when an event is emitted from within the Chart. However, SIMULINK can trigger the Chart in certain cases. These are the following

- When SIMULINK decides to sample the STATEFLOW Chart block, according to the sampling times of the input signals. Indeed STATEFLOW can have inputs from the SIMULINK model that have sample time, according to which the Chart is updated. Also we can set the sample time parameter of the Chart into a certain period as well.
- A STATEFLOW block can accept a trigger, pretty much like a SIMULINK subsystem can be triggered by a signal. Thus we can declare a Chart input as a trigger, which, as in SIMULINK, will trigger the Chart depending on the *rising, falling* or *either* choice we make. See Section 4.5.2 for a description of different types of SIMULINK triggers.

## 5.2 Semantical issues with STATEFLOW

One of the motivations for this work was to define explicitly a "safe part" STATEFLOW and then provide means for checking and possibly correcting

#### Non-termination and stack overflow

As already mentioned, a transition in STATEFLOW can be multi-segment and the segment graph can have cycles. Such a cycle can lead to non-termination of the interpretation algorithm during the search for valid transition step.

Another source of potential problems is the run-to-completion semantics of event broadcast. Every time an event is emitted the interpretation algorithm is called recursively, runs to completion, then execution resumes from the action statement immediately after the emission of the event. This can lead, semantically, to infinite recursion and in practice (i.e., during simulation) to stack overflow.[4]

A simple model resulting in stack overflow is shown in Figure 5.1. When the default state $A$ is entered, the event $E$ is emitted as instructed by the entry action of $A$. $E$ results in a recursive call of the interpretation algorithm and since $A$ is active its outgoing transition is tested. Since the current event $E$ matches the transition event (and because of the absence of condition) the condition action is executed, emitting $E$ again. This results in a new call of the interpretation algorithm which repeats the same sequence of steps filling up the stack until overflow.

[4] This is recognized in the official documentation: *"Broadcasting an event in the action language is most useful as a means of synchronization among AND (parallel) states. Recursive event broadcasts can lead to definition of cyclic behavior. Cyclic behavior can be detected only during simulation."*

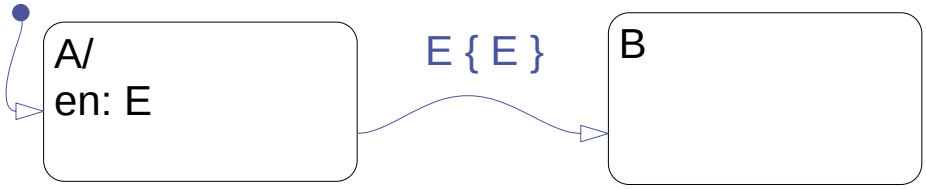



**Figure 5.1:** Stack overflow

### Backtracking without "undo"

While searching for a valid transition, STATEFLOW explores the segment/junction graph, until a destination state is reached. If, during this search, a junction is reached without any enabled outgoing segments, the search backtracks to the previous junction (or state) and looks for another segment. This backtrack, however, does not restore the values of variables which might have been modified by a condition action. Thus, the search for valid transitions can have side effects on the values of variables.

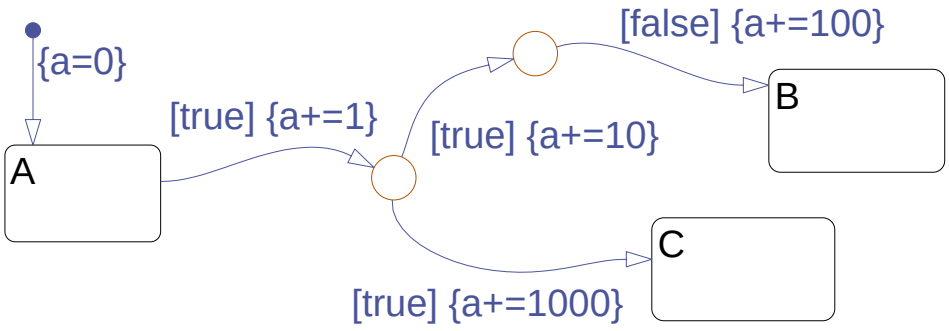



**Figure 5.2:** Example of backtracking

An example of such a behavior is generated by the model shown in Figure 5.2. The final value of variable $a$ when state $C$ is entered will be 1011 and not 1001 as might be expected. This is because when the segment with condition "false" is reached, the algorithm backtracks without "undoing" the action "a+=10".

### Dependence of semantics on graphical layout

In order to enforce determinism in the search order for active states and valid transitions (thus ensuring that the interpretation algorithm is deterministic) STATEFLOW uses two rules: the "top-to-bottom, left-to-right" rule for states and the "12 o'clock" rule for transitions. These rules imply that the semantics of a model depend on its graphical layout. For example, as the model

is drawn in Figure 5.3, parallel state $A$ will be explored before $B$ because it is to its left. But if $B$ was drawn slightly higher, then it would be explored first. (Notice that STATEFLOW annotates parallel states with numbers indicating their execution order, e.g., as shown in Figure 5.3.)

The order of exploration is important since it may lead to different results. In the case of "12 o'clock" rule, for example, if the top-most transition of the model of Figure 5.2 emanated from the 11 o'clock position instead of the 1 o'clock position, then the final value of $a$ would be 1001 instead of 1011.

Exploration order also influences the semantics in the case of parallel states, even in the absence of variables and assignments. An example is given by the model of Figure 5.3. $A$ and $B$ are parallel states. When event $E_1$ arrives, if $A$ is explored first, then $E_2$ will be emitted and the final global state will be $(A_2, B_3)$. But if $B$ is explored first then state $B_2$ is reached and, when exploring state $A$ the emitted event $E_2$ will not change the state of $B$, since from state $B_2$ there is no output transition. Thus, the final global state will be $(A_2, B_2)$. This means that parallel states in STATEFLOW do not enjoy the property of *confluence*.

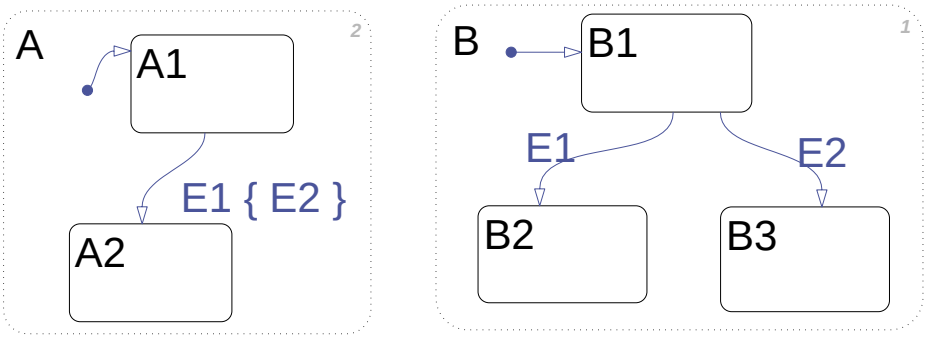



**Figure 5.3:** Example of non-confluence

### Early return logic

Another interesting feature of STATEFLOW is termed *early return logic* in the STATEFLOW manual. This problem is illustrated in Figure 5.4. When event $E$ is emitted, the interpretation algorithm is called recursively. Parent state $A$ is active, thus, its outgoing transition is explored and, since event $E$ is present, the transition is taken. This makes $A$ inactive, and $B$ active. When the stack is popped and execution of the previous instance of the interpretation algorithm resumes, state $A_1$ is not active anymore, since its parent is no longer active.

The problem with early return logic may arise when the user has a program, in the STATEFLOW action language, where there are variable manipulation and event emits. If an event emitted will cause an early return, the user may have the false impressions that the entire body of his program is executed, updating all the variables.

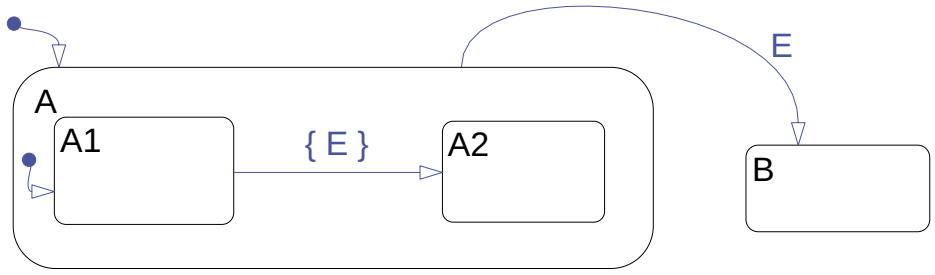



**Figure 5.4:** "Early return logic" problem

### Super-transitions

We end this part by briefly mentioning a last problem; the possibility of having the so-called *super-transitions* crossing different levels of the state hierarchy. This is a feature of Statecharts as well, but is generally considered harmful in the Statecharts community [Har87]. Many proposals disallow such transitions for the sake of simpler semantics [LvdBC00].

## 5.3 Simple conditions identifying a "safe" subset of STATEFLOW

In this section we present a sufficient number of simple conditions for avoiding error-prone models such as those discussed previously. The conditions can be statically checked using mostly light-weight techniques. The conditions identify a preliminary, albeit strict, "safe" subset of STATEFLOW. A larger subset can be identified through "heavier" checks such as model-checking, as discussed in Section 5.6.

**Absence of multi-segment loops:** If no graph of junctions and transition segments contains a loop (a condition which can easily be checked statically) then the model will not suffer from non-termination problems referred to in Section 5.2. This condition is quite strict, however, it is hard to loosen, since termination is undecidable for programs with counters and loops.

**Acyclicity in the graph of triggering and emitted events:** An event $E$ is said to be *triggering* a state $s$ if the state has a "on event $E$: $A$" action or an outgoing transition which can be triggered by $E$ (i.e., $E$ appears in the event field of the transition label or the event field is empty). $E$ is said to be *emitted* in $s$ if it appears in the entry, during, exit or on-event action of $s$, or in the condition or transition action[5] of one of the outgoing transitions of $s$. Given a STATEFLOW model, we construct the following graph. Nodes of the graph are all states in the model. For each pair of nodes $v$ and $v'$, we add an edge $v \rightarrow v'$ iff the following two conditions hold:

[5] In fact, transition action events can probably be omitted from the set of emitted events of $s$, resulting in a less strict check. We are currently investigating the correctness of this modification.

1. There is an event $E$ which is emitted in $v$ and triggering $v'$.
2. Either $v = v'$ or the first common parent state of $v$ and $v'$ is a parallel state.

The idea is that $v$ can emit event $E$ which can then trigger $v'$, but only if $v$ and $v'$ can be active at the same time. If the graph above has no directed cycle then the model will not suffer from stack overflow problems.

**Absence of assignments in intermediate segments:** In order to avoid side effects due to lack of "undo", we can simply check that all variable assignments in a multi-segment transition appear either in transition actions (which are executed only once a destination state has been reached) or in the condition action of the last segment (whose destination is a state and not a junction). This ensures that even in case the algorithm backtracks, no variable has been modified. An alternative is to avoid backtracking altogether, as is done with the following check.

**Conditions of outgoing junction segments form a cover:** In order to ensure absence of backtracking when multi-segment transitions are explored, we can check that for each junction, the disjunction of all conditions in outgoing segments is the condition *true*. If segments also carry triggering events, we must ensure that all possible emitted events are covered as well.

**Conditions of outgoing junction segments are disjoint:** In order to ensure that the STATEFLOW model does not depend on the 12 o'clock rule, we must check that for each state or junction, the conditions of its outgoing transitions are pair-wise disjoint. This implies at most one transition is enabled at any given time. In the presence of triggering events, we can relax this by performing the check for each group of transitions associated with a single event $E$ (or having no triggering event).

It should be noted that checking whether STATEFLOW conditions are disjoint or form a cover is an undecidable problem, because of the generality of these conditions. From a STATEFLOW design, we can extract very easily the logical properties expressing that a set of conditions are disjoint and form a cover. These logical properties can be transmitted as a proof obligation to some external tool such as a theorem prover. However, for most practical cases, recognizing common sub-expressions is sufficient for establishing that some conditions are disjoint and form a cover.

**Checks for confluence:** In order to ensure that the semantics of a given STATEFLOW model does not depend on the order of exploring two parallel states $A$ and $B$, we must check two things. First, that $A$ and $B$ do not access the same variable $x$ (both write $x$ or one reads and the other writes $x$). But this is not sufficient, as shown in Section 5.2, because event broadcasting alone can cause problems. A simple solution is to check that in the aforementioned graph of triggering and emitted events, there is no edge $v \rightarrow v'$ such that $v$ belongs to $A$ and $v'$ to $B$ or vice-versa.

This is exactly the problem of Figure 5.3, where one of the edges in the above mentioned graph of triggering and emitted events is the $A_1 \rightarrow B_1$; the emitted event in the outgoing transition form $A_1$ to $A_2$ is the same event $E$ that is triggering event for the state $B_1$ (because the

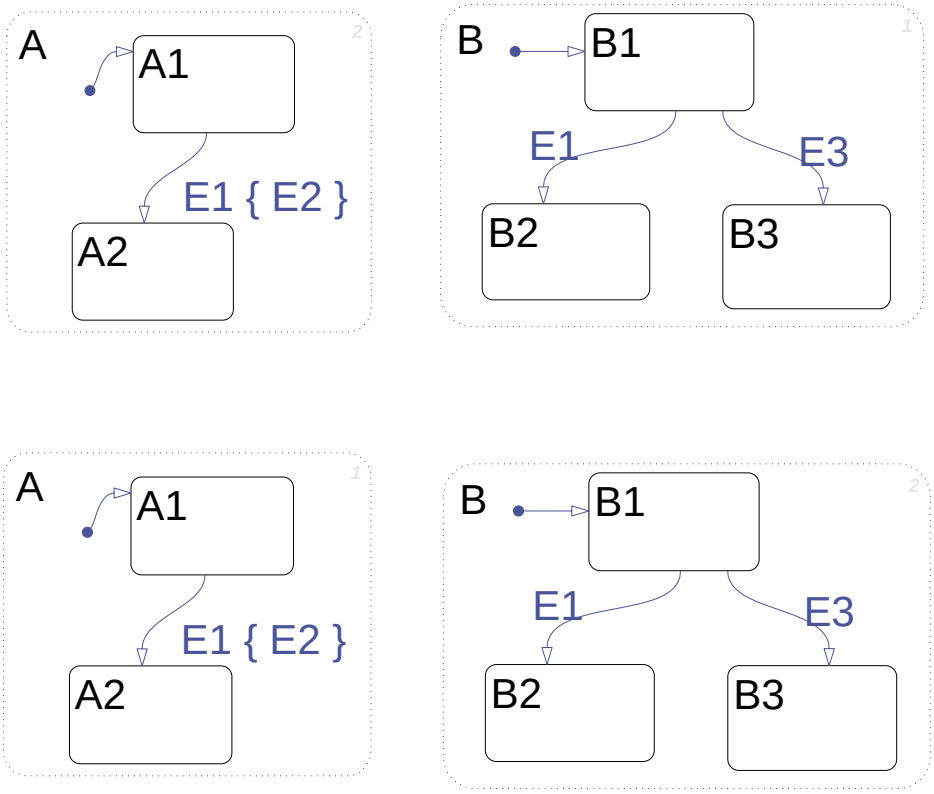



**Figure 5.5:** The example of Figure 5.3 corrected to be confluent

outgoing transition to state $B_3$ is triggered by $E$). Also the first common parent of states $A_1$ and $B_1$ is a parallel state (the root).

Thus if the triggering event in the outgoing transition of state $B_1$ to $B_3$ is another event, i.e., $E_3$, the design would be confluent. This is the case of the designs in Figure 5.5, where we substituted the $E_2$ event with $E_3$ not emitted in the parallel state $A$ (or in any sub-state). Note that the both designs in Figure 5.5 are the same with the only difference the graphical layout; in the bottom design, state $B$ is below state $A$ and it will be executed/checked after that. And in the upper design it is the contrary (we can see that also by the numbers in the upper right corner of each state). In any case, as said earlier, both designs are confluent and will have the same effects no matter the input.

**Checks for "early return logic":** To ensure that our model is free of "early return logic" problems, we can check that for every state $s$ and each of its outgoing transitions having a triggering event, this event is not emitted anywhere inside $s$ or its eventual sub-states. Note that if a transition has no triggering event then this transition is enabled for any event, thus, we must check that no event is emitted in $s$.

## 5.4 Translation into LUSTRE

The checks on a STATEFLOW model described in Section 5.3 define a subset which is much more likely to be correct according to the system designer's intentions than using the full STATEFLOW definition. It is restrictive, however, since it disallows some of STATEFLOW's programming features which designers have become used to. We would therefore like to extend our subset by employing analysis with sound theoretical underpinnings. One such framework is model-checking and we have access to the well-established model-checker called LESAR [RHR91, HLR92] which takes LUSTRE as its input. A translation of STATEFLOW into LUSTRE therefore opens up the possibility of allowing some of the "unsafe" features of STATEFLOW to be used with confidence provided we can verify the intended properties of the model using LESAR.

We have to be clear, however, about the difference between the subset of STATEFLOW which is "safe" in the sense of the previous discussion and that which is translatable into LUSTRE. We can copy the behavior of STATEFLOW as precisely as required (given sufficient effort in building the translator) and can even implement loops and recursion *provided we can prove that the behavior is bounded*. The generated program, however, does not have any guaranteed safety properties since all the previous discussion about the semantical problems with STATEFLOW are carried over into the LUSTRE translation. This is where model-checking and other formal methods can be applied. In this section we describe the translation process informally and in Section 5.6 we show how some of the previously mentioned properties can be verified and our subset extended using the LESAR model-checker.

Needless to say, the goal of the translation is not simply to provide a way to model-check STATEFLOW models. It is also to allow for semantic-preserving code generation and implementation on uni-processor or multi-processor architectures [CCM+03, TSSC05, STC05].

### 5.4.1 Encoding of states

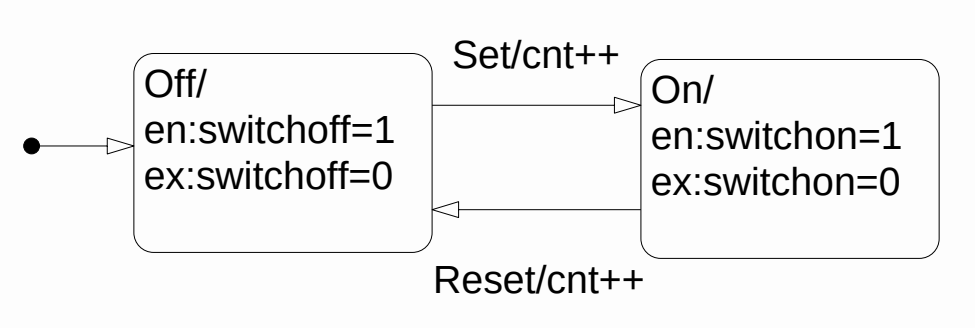



**Figure 5.6:** A simple STATEFLOW chart

The most obvious method of encoding states into LUSTRE is to represent each state as a boolean variable and a section of code to update that variable according to the validity of the

input and output transitions. For example, one can envisage a very simple and elegant encoding of the boolean component (i.e., *without* the entry actions) of the example in Figure 5.6 in the LUSTRE code depicted in Figure 5.7. Here a state becomes true if any of its predecessor states are true and there is a valid transition chain from that state. It becomes false if it is currently true and there is a valid transition chain to any of its successor states. Otherwise it remains in the same state. The initial values of the states are defined by the validity of the default transitions.

```
node SetReset0(Set, Reset: bool)
returns (sOff, sOn: bool);
let
  sOff = true ->
    if pre sOff and Set then false
    else if (pre sOn and Reset) then true
    else pre sOff;
  sOn = false ->
    if pre sOn and Reset then false
    else if (pre sOff and Set) then true
    else pre sOn;
tel
```

**Figure 5.7:** Simple LUSTRE encoding of the example

This code is semantically correct for a system consisting only of states but it is difficult to incorporate the imperative actions attached to both states and transitions in STATEFLOW. For example, if the above code had included the entry actions in the states then all the values referenced by the action code would have to be updated in each branch of the if-then tree. This causes two problems. Firstly, for even quite small charts the number of values being updated can become large and this has to be multiplied by the complexity introduced by the network of transitions each state participates in. Secondly, the action language is an imperative language for which it would be difficult to compile a single expression for each sequence of actions. Note also that if more than one state updates the same value then causality loops and multiple definitions could arise.

A more practical approach, therefore, is to split the above equations into their components and use explicit dependencies to force their order of evaluation. Inspecting the code in Figure 5.7 the state update equation for each state consists of:

- an initialization value computed from default transitions (`true` for `sOff`),
- a value for each outgoing transition (`Set` for `sOff`),
- an exit clause (`(pre sOff and Set)` for `sOff`),
- an entry clause (`(pre sOn and Reset)` for `sOff`) and
- a no-change value (`pre sOff`).

Explicitly separating these components allows us to insert the action code at the correct point in the computation of a reaction. This results in the rather dense encoding shown in Figure 5.8. Here, the code has been split into several sections.

- **Initial values.** These are the initial values for all variables, `false` for states and the initial value from the data dictionary for STATEFLOW variables. Mention that after the first, initial cycle, those variables have the `pre` value of the corresponding variable, which is the previous state (on or off) of the corresponding state (in the STATEFLOW graph)
- **Transition validity.** In this section the values for the transitions are computed. For convenience in the translator these are actually calls to predefined nodes generated in advance from the transitions' events and actions. Note that the test for the activity of the source state is included in the transition's validity test. In the example of Figure 5.8, for reasons of simplicity, the validity of the links appears inside the node.
- **State exits.** Any states which are `true` and have a valid outgoing transition are set to `false`.
- **Exit actions.** The code for any exiting state's exit actions is computed. This section also includes during actions for states which remain active and `on` actions also for active states.
- **Transition actions.** The code for the transition actions is executed. Note that the exiting state's value is `false` while this occurs.
- **State entries.** Any states which are `false` and have a valid incoming transition are set to `true`.
- **Entry actions.** Entering states action code is executed with the state's variable now `true`.

This sequence corresponds to the sequence of events in STATEFLOW's interpretation algorithm. Note that by "transition valid" we do not mean that the transition is valid with respect to the current context but that this is a transition which will be traversed in the current reaction. Thus the arbitration between competing outgoing transitions has to be resolved by the transition valid computation.

There are some additional complications in the code shown in Figure 5.8, for instance the use of the `init` and `term` flags which are used to control initialization and termination of subgraphs but these are discussed in the later sections.

### 5.4.2 Compiling transition networks

Figure 5.9 shows a STATEFLOW chart with a junction. Junctions in STATEFLOW do not have a physical state and can be thought of as nodes in an `if-then` tree. This is thus the most sensible encoding of junctions. One problem, however, is that junction networks can be sourced from more than one state and/or a single state can have more than one output to the same junction.

```
node SetReset1(Set, Reset, init, term: bool)
returns (sOff, sOn: bool; switchon, switchoff, cnt: int);
var sOff_1, sOff_2, sOn_1, sOn_2, lv5, lv6, lv7: bool;
    switchon_1, switchon_2, switchoff_1, switchoff_2,
    cnt_1, cnt_2: int;
let
  -- initial values
  sOff_1      = false -> pre sOff;
  sOn_1       = false -> pre sOn;
  switchon_1  = 0     -> pre switchon;
  switchoff_1 = 0     -> pre switchoff;
  cnt_1       = 0     -> pre cnt;
  -- link validity
  lv5 = if sOff_1 then Set   else false;
  lv6 = if sOn_1  then Reset else false;
  lv7 = if init and not (sOff_1 or sOn_1) then true else false;
  -- state exits
  sOff_2 = if sOff_1 and (lv5 or term) then false else sOff_1;
  sOn_2  = if sOn_1  and (lv6 or term) then false else sOn_1;
  -- exit actions
  switchoff_2 = if not sOff and sOff_1 then 0 else switchoff_1;
  switchon_2  = if not sOn  and sOn_1  then 0 else switchon_1;
  -- transition actions
  cnt_2 = if lv5 then cnt_1+1 else cnt_1;
  cnt   = if lv6 then cnt_2+1 else cnt_2;
  -- state entries
  sOff = if not sOff_2 and (lv7 or lv6) then true else sOff_2;
  sOn  = if not sOn_2  and lv5 then true else sOn_2;
  -- entry actions
  switchoff = if sOff and not sOff_1 then 1 else switchoff_2;
  switchon  = if sOn  and not sOn_1  then 1 else switchon_2;
tel
```

**Figure 5.8:** Alternative LUSTRE encoding of the example

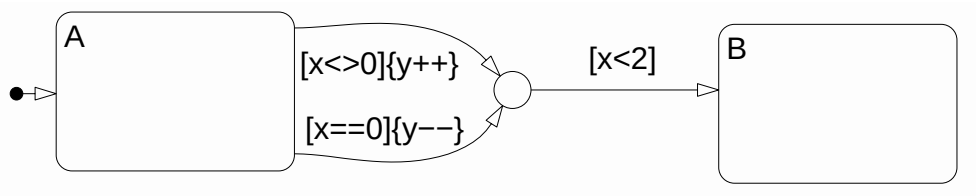



**Figure 5.9:** A STATEFLOW chart with a junction

These can be handled quite easily if one allows a certain amount of code duplication, the common subnetwork for two joining outgoing transitions being compiled twice.

We could devise a very natural scheme for LUSTRE to handle this but again it becomes difficult to insert the condition and transition actions into the `if-then` tree in LUSTRE. Figure 5.10 shows the actual code generated[6]. The functions `cv{678}_` not shown compute the condition code for their respective transitions. Note how the `cv8_` call is duplicated between `lv6_` and `lv7_`. Essentially, the junction tree is turned into a flattened representation with two flags, "`end`" which signifies the termination of the tree (either a destination state or a terminal junction) and "`exit`" which is true if the terminal was a state. One slight inefficiency is the use of these flags to defeat further computation after the termination point is reached (the `not end` clauses). These two flags correspond to the `End`, `No` and `Fire` transition values in [HR04], the semantics of our junction processing is identical to the semantics described therein.

There is also a slight problem with the "transition valid" section in the code shown in Figure 5.8. For the example shown there can only ever be one transition valid flag `true` at each instant but when a state has (potentially competing) outgoing transitions there has to be some kind of arbitration between them, hopefully using the same arbitration as STATEFLOW itself. In fact the statements are chained together with a common flag which indicates when a valid transition has been found. This is called the `ok` variable and a revised transition validity computation section is shown in Figure 5.11. In fact we need a separate `ok` flag *for each subgraph*, this is explained later when inter-level transitions are discussed.

A more serious problem is that junction networks can have loops which results in unbounded recursion and therefore a loss of synchronous semantics. Figure 5.12 shows a simple `for`-loop. There are a number of possibilities for handling this.

**Junctions as states.** An easy solution would be to give junctions a physical state in the executing LUSTRE program. This effectively moves the non-termination problem outward into the code calling the STATEFLOW model but also moves the burden of the proof of non-termination to the client code. This has been implemented in our translator where we also provide an additional status flag called "`valid`" as an output which is `true` only if the current state is not a junction. In theory, the client code could loop over the STATEFLOW code until this flag becomes `true` at which point the other outputs are also valid.

**Loop unrolling with external proof obligations.** This is unsatisfactory from the point of view of using the translator as a development tool. We would prefer to simply impose a synchronous semantics upon STATEFLOW and outlaw such constructs if they cannot be proven to be bounded. Given a synchronous semantics for STATEFLOW we have to outlaw such constructs in the general case. It is possible, however, to unroll such loops (Figure 5.12 also shows the expansion of the simple loop) without loss of generality, provided bounds can be proven on the number of iterations. This means we can generate proof obligations for external tools such as Nbac [JHR99]. If

[6] Our code examples have been condensed for brevity and use abbreviated variable names. `cv` means "condition valid", `lv` "transition valid", `ca` "condition action" and `su` "state update"

```
-- link id=7 name=[x<>0]{y++}
node lv7_(x, y: int; ok, lv7, lv8: bool)
returns(yo: int; oko, lv7o, lv8o: bool);
var cv7, cv8, end, end_1, end_2, ok_1: bool;
let
  end_1 = false;
  ok_1, cv7, end_2 =
    if (not (end_1 or ok)) then cv7_(x) else (ok, false, end_1);
  yo = if cv7 then ca7(y) else (y);
  oko, cv8, end =
    if ((not end_2) and cv7) then cv8_(x) else (ok_1, false, end_2);
  lv7o, lv8o = if (cv8 and end) then (true, true) else (lv7, lv8);
tel


-- link id=6 name=[x==0]{y--}
node lv6_(x, y: int; ok, lv6, lv8: bool)
returns(yo: int; oko, lv6o, lv8o: bool);
var cv6, cv8, end, end_1, end_2, ok_1: bool;
let
  end_1 = false;
  ok_1, cv6, end_2 =
    if (not (end_1 or ok)) then cv6_(x) else (ok, false, end_1);
  yo = if cv6 then ca6(y) else (y);
  oko, cv8, end =
    if ((not end_2) and cv6) then cv8_(x) else (ok_1, false, end_2);
  lv6o, lv8o = if (cv8 and end) then (true, true) else (lv6, lv8);
tel


-- node id=3 name=A
node suAlv(x, y: int; ok, sA, trm, ini: bool)
returns(yo: int; oko, lv6, lv7, lv8: bool);
var lv8_1, ok_1: bool; y_1: int;
let
  y_1, ok_1, lv6, lv8_1 = lv6_(x, y, ok, false, false);
  yo, oko, lv7, lv8 = lv7_(x, y_1, ok_1, false, lv8_1);
tel
```

**Figure 5.10:** Code generated for the junctions example

```
ok_1 = false;
lv5, ok_2 = if not ok_1 and sOff_1
            then (Set, Set) else (false, ok_1);
lv6, ok_3 = if not ok_2 and sOn_1
            then (Reset, Reset) else (false, ok_2);
lv7, ok   = if not ok_3 and init and not (sOff_1 or sOn_1)
            then (init, init) else (false, ok_3);
```

**Figure 5.11:** Chaining together transition-valid computations

a bound exists and is feasible we can unroll loops individually as required. This requires further investigation. Currently, we detect all junction loops and reject models which have them.

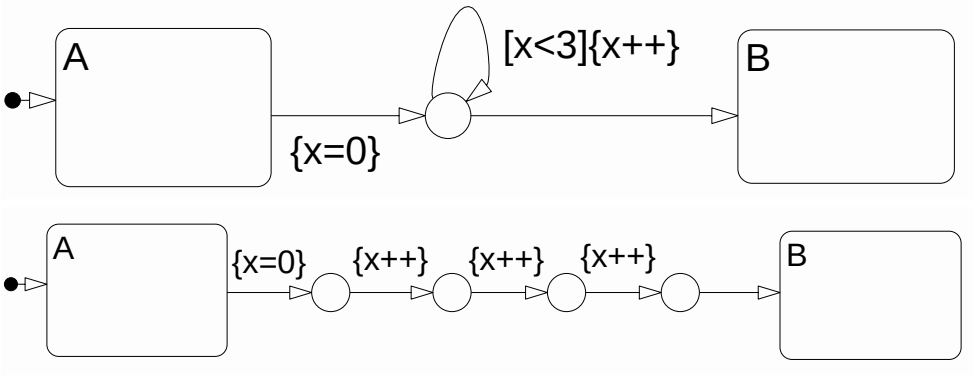



**Figure 5.12:** A `for`-loop implemented in STATEFLOW junctions and its expansion

### 5.4.3 Hierarchy and parallel AND states

The entire hierarchy of a STATEFLOW model and the translation of it towards LUSTRE nodes boils down to simple function calls of nested states, the only complication being the initialization and termination of the nested states.

For example, Figure 5.13 illustrates a simple model with both parallel and exclusive substates. For both types of substate we insert the function calls to the substates after computation of the local state variables, the LUSTRE nodes generated for the top-level state (parallel) and state `B` (exclusive) for this model are depicted in Figure 5.14.

Initialization and termination are controlled by two variables, "`ini`" and "`trm`" which are passed down the hierarchy. This is a standard method for implementing state machines in synchronous languages [MH96]. One way of viewing the `ini` value is as a*pseudo-state* which the model is in prior to execution and in fact this plays the role of the state variable for default transitions. For parallel states the local state variable depends only on the `ini` and `trm` variables, as do the flags for entry, exit and during actions. These are computed as in Figure 5.15 (`s` is

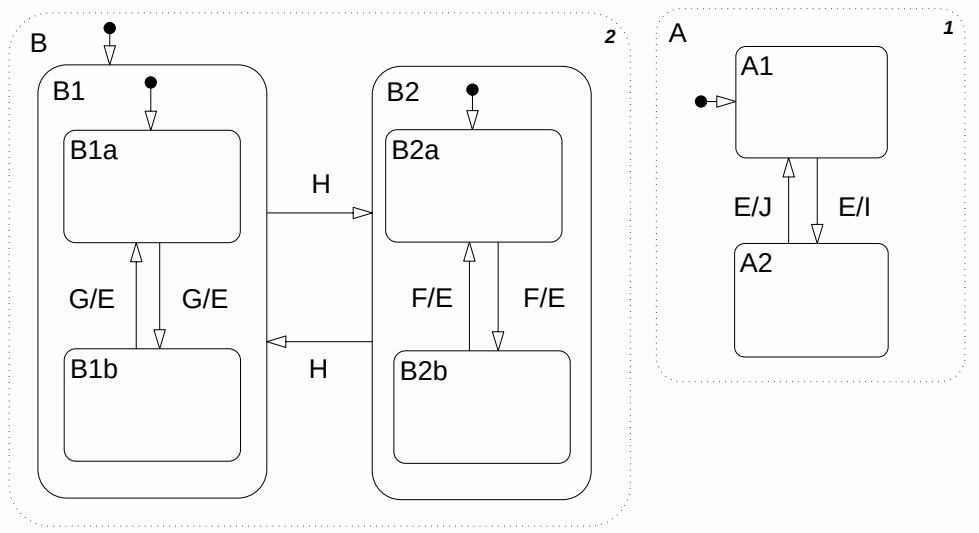



**Figure 5.13:** A model with parallel (AND) and exclusive (OR) decompositions

the local state variable) and are embodied in auxiliary nodes (for example the state variable is computed by the node `sfs` in Figure 5.14).

For exclusive substates the `ini` and `trm` flags are computed solely from the local state variable (`ini = s and not pre s` and `trm = pre s and not s`). The complication is that we need the value of the state variable at the end of the reaction without actually setting the variable itself because the nested states have to be executed using the input value. This is why we call the state entry computation beforehand (`sgu8_B1en` for example) but save the value in a temporary variable (`sg8_B1t`) and then update the actual value at the end of the computation. The temporary value then stands for the new value and the input value (`sg_8B1in`) for the previous one. Actually, for the code presented here this is unnecessary but when event broadcasting is enabled (Section 5.4.6) the value of the state variable can be updated by actions. Note also that for the top-level call we set `ini` to `true -> false` and `trm` to `false`.

### 5.4.4 Inter-level and inner transitions

The methods described so far work in a natural way for STATEFLOW charts which are structured as trees, which allows the LUSTRE code also to be structured as a tree. One consequence of this is that we can map states onto LUSTRE nodes and still retain the same action sequences as STATEFLOW. STATEFLOW, however, allows inter-level transitions, i.e., between states not at the same level of the node hierarchy which means that the model becomes a more general graph structure rather than a tree. This in itself does not break any of the characteristics of a synchronous implementation but it does greatly complicate the translation. As such, early versions of the translator simply outlawed transitions of this type in favor of a much simpler analysis. A large amount of legacy STATEFLOW code uses inter-level transitions, however, so a

```
-- Toplevel graph (AND,[A,B])
node sf_2(F,G,H: event) returns(I,J: event);
let
  ...
  sgA = sfs(ini,trm);
  J,I,okA,sA1,sA2 = sf_4(E_1,I_1,J_1,okA_1,sA1_1,sA2_1,sgA,trm,ini);
  sgB = sfs(ini,trm);
  okB2,okB1,okSubgraph36,sB2a,sB2b,sB1a,sB1b,sgB2,sgB1,E =
    sf_7(F,G,H,E_1,okB2_1,okB1_1,okSubgraph36_1,sB2a_1,sB2b_1,sB1a_1,
         sB1b_1,sgB2_1,sgB1_1,sgB,trm,ini);
  ...
tel
```

```
-- State B (OR,[B1,B2])
node sf_7(F,G,H,E: event; okB2,okB1,okSubgraph36,sB2ain,sB2bin,sB1ain,
          sB1bin,sgB2in,sgB1in,sgB,trm, ini: bool)
returns(okB2o,okB1o,okSubgraph36o,sB2a,sB2b,sB1a,sB1b,sgB2,sgB1: bool;
        Eo: event);
let
  ...
  sgB1t = sguB1en(okSubgraph36o,lv16,lv18,sgB1_1,trm,ini);
  sgB2t = sguB2en(okSubgraph36o,lv17,sgB2_1,trm,ini);
  okB1o,sB1a,sB1b,E_1 =
    sf_8(G,E,okB1,okSubgraph36o,lv17,sB1ain,sB1bin,sgB1t,
         ((not sgB1t) and sgB1in),(sgB1t and (not sgB1in)));
  okB2o,sB2a,sB2b,Eo =
    sf_11(F,E_1,okB2,okSubgraph36o,lv18,sB2ain,sB2bin,sgB2t,
          ((not sgB2t) and sgB2in),(sgB2t and (not sgB2in)));
  ...
tel
```

**Figure 5.14:** LUSTRE code fragments for parallel and hierarchical states

| | |
|---|---|
| state | `(init and not term) ->`<br>`(init or pre s) and (not term)` |
| entry | `init -> s and not pre s` |
| exit | `(init and term) ->`<br>`((pre s or ((not pre s) and init)) and (not s))` |
| during | `false -> s and pre s` |

**Figure 5.15:** Computation of parallel state variables

preliminary version of our translator which can handle inter-level transitions has been developed.

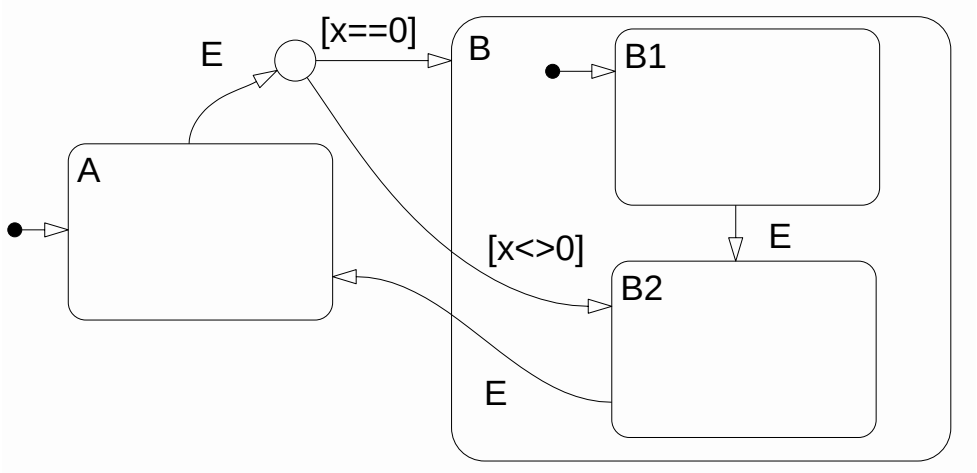



**Figure 5.16:** A model with inter-level transitions

Figure 5.16 illustrates a simple STATEFLOW chart with an inter-level transition network, from `A` to `B` and `B2`. Figure 5.17 shows the different kinds of inner transitions that can be used. The top transition `[x==0]` is an inner transition which terminates in the parent state `A`, transitions `[x==1]` and `[x==2]` show inner transitions to and from a substate of `A` and transition `[x==3]` terminates in a junction (this style of inner transition is known as a *flowchart* in STATEFLOW terminology).

These charts show a number of problems with inter-level transitions:

- The inter-level transition from the junction to `B2` in Figure 5.16 acts *in lieu* of a default transition when it is taken so any default transition in the states traversed by the path have to be ignored.
- Any transition which traverses a state inwards results in activation of that state and likewise any transition which traverses outwards results in deactivation of the state.

and inner transitions:

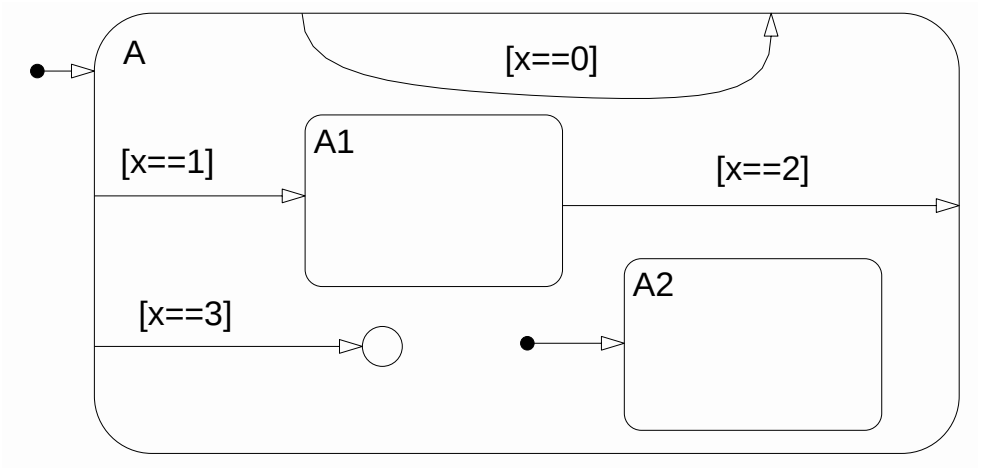



**Figure 5.17:** A model with inner transitions

- The semantics of inner transitions mean that, for example, transition `[x==1]` in Figure 5.17 acts as a default transition for state `A` *when an inner transition results in termination of the currently active substate but not when the state is entered from outside*. The other three transitions do not have this property since none of them terminate in an internal state.
- State `A` neither exits nor enters when an inner transition is taken and its `during` actions are executed *before* the inner transitions are taken. Thus, if either of the transitions `[x==0]` or `[x==2]` are taken state `A2` is reached.
- Note that transitions `[x==0]`, `[x==1]` and `[x==3]` are considered to emanate from the same source and thus require arbitration and are subject to STATEFLOW's check for multiple valid transitions. They also take precedence over default transitions when an inner transition is taken.
- Only one inner transition can be taken at a time so that if state `A1` exits on transition `[x==2]` it cannot return on transition `[x==1]`.
- Flowchart transitions are taken, if valid, each time the state is active and a higher priority transition is not valid. Inner transitions are prioritized according to the 12 o'clock rule so that they are checked in the order `[x==0]`, `[x==3]` then `[x==1]`. Inner transitions from the parent state take precedence over those emanating from substates. They do not result in a change of state and are evaluated purely for their side-effects.

In addition, a transition network can be *mixed*, i.e., has paths through it which can be inter-level, inner, flowchart or normal paths, or an arbitrary combination of all of them. Note also that both inner and inter-level transitions can lead to inconsistent states if not implemented properly. This results in a highly complex semantics for STATEFLOW transitions which would be extremely difficult to emulate precisely. The semantics in [HR04] follows STATEFLOW's interpretation

algorithm very closely but is essentially an imperative method which would be difficult to adapt to LUSTRE's synchronous semantics.

We have, instead, implemented a compromise solution which behaves in a very similar manner to STATEFLOW with some distortions on the state, condition and transition actions. This solution is based on splitting transition networks into separate *paths* and associating them with the *outermost* point traversed by any transition in the path. Evaluation then proceeds top-down as before but computing transition validity is done when the transitions come into scope. The results of this computation can then be passed down the hierarchy. For instance, the transitions labeled `E` and `F` in Figure 5.18 are computed at the top level of the hierarchy and then flags corresponding to their validity are passed as arguments to the nodes generated for states `A` and `B`. States `A1` and `B1` then include these additional parameters in their entry and exit clauses.

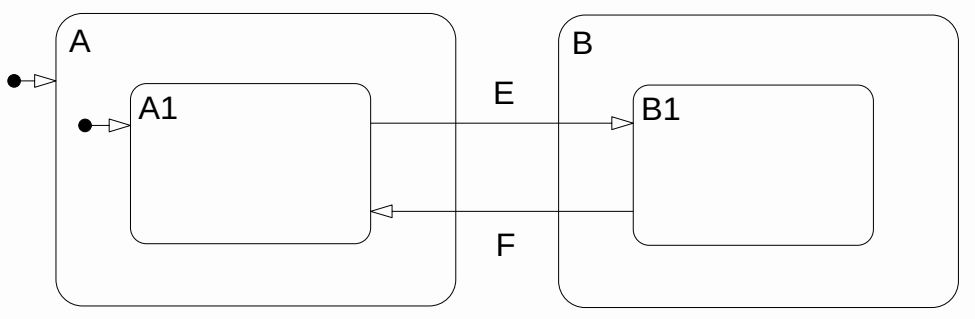



**Figure 5.18:** Inter-level transitions with action order distortion

The problem then arises as to how to ensure that the sequence of state exit action followed by transition action followed by state entry action is in the correct order. If substates are checked in a fixed order then at least one of transitions `E` or `F` must be evaluated in reverse, i.e., the entry and exit actions will be executed in the wrong order. Several possible solutions are possible:

- We could dynamically order the calls to the nodes for `A` and `B` according to which transitions have been computed as valid.
- We could move the entry, transition and exit actions to either the source for the transition path, the outermost scope of the transition path or the destination of the transition path.
- We could lift all the actions to the top-level node in the hierarchy and impose an order on the actions based upon some abstraction of STATEFLOW's interpretation algorithm.

All of these options would result in other more subtle distortions in the actions as compared to STATEFLOW. They also have the additional complexity of computing all the entry and exit actions for states along the paths traversed.

Currently, none of these options are implemented so we can only guarantee the correct order of action execution only for *for non-inter-level transitions*. We can, however, guarantee that all actions which would have been executed within a single LUSTRE reaction will get executed in some order.

#### Inter-level transitions

The basic scheme, however, is relatively easy to implement for inter-level transitions provided we are careful to compute the correct arguments (transitions) to the substate nodes.

The only major complication is the computation of the `ok` value for inter-level transitions. Because of the presence of default transitions we need an `ok` flag for each substate because the computation of transition validity is disjoint for each default transition taken within the hierarchy. We also need a separate `ok` flag for each parallel state because transition computations are also disjoint between parallel states. Luckily, STATEFLOW outlaws inter-level transitions between parallel states but we still need a flag for each subgraph because of default transitions. This means that we need to associate an `ok` value with each transition (in fact we associate it with the flag for its *source* graph) so that the transition is only valid if both its validity flag and associated `ok` flag are true.

Figure 5.19 shows the code produced for Figure 5.16. This is a direct implementation of the scheme described above. Points to note about this code include:

- transitions 8 (`default for A`), 9 (`A to junction`), 10 (`B2 to A`), 12 (`junction to B`) and 13 (`junction to B2`) are all computed at the top-level, of which 9, 10 and 13 are passed to the node for subgraph `B`,
- the node for state `B` augments these with the transitions 11 (`B1 to B2`) and 14 (`default for B1`),
- the complex predicate for the default transition to state `B1` has to take into account whether state `B` is being entered by inter-level transition 13 or normal transition 12,
- the distortions in the actions (state `A` enters before `B2` exits if transition 10 is taken) and
- the computation of the `ok` flags, for example, transition 12 uses `okTop` whereas transition 10 uses `okB`.

#### Inner transitions

Although we have treated inter-level and inner transitions separately here, they are intimately interlinked due to the possibility of a single path through a transition network having transitions of both types. It is even possible for a single transition to be of both types. In some ways, inner transitions are simpler than inter-level transitions since they are nearly local (only involving the immediate parent state) but are more complicated in the way they interact with other transitions at the same level.

### 5.4.5 Action language translation

There are two basic options for translating the simple imperative language implemented by STATEFLOW into the synchronous language LUSTRE. One possibility would be to generate C code from the action code and use the external function call facility of LUSTRE to call the action

```
-- link id=8  name=        point -> A
-- link id=9  name=E       A -> junction
-- link id=10 name=E       B2 -> A
-- link id=11 name=E       B1 -> B2
-- link id=12 name=[x==0]  junction -> B
-- link id=13 name=[x<>0]  junction -> B2
-- link id=14 name=        point -> B1
-- graph id=17 name=B,NONTOP
node sf_4(E:event; okB,okTop,lv9,lv10,lv13,sB1in,
          sB2in,sgB,trm,ini:bool)
returns(okBo,sB1,sB2:bool);
var lv11,lv11_1,lv14,lv14_1,okB_1,sB1_1,sB2_1:bool;
let
  lv11_1,lv14_1=(false,false);
  okB_1,lv11=if sB1in then suB1lv(E,okB,sB1in,trm,ini)
             else (okB,lv11_1);
  okBo,lv14=if not ((okTop and lv13) and (okTop and lv9))) and
               (ini and (not (sB2in or sB1in))
            then iniu19__pointlv(okB_1,trm,ini) else (okB_1,lv14_1);
  sB2_1=if sB2in then suB2ex(okBo,lv10,sB2in,trm,ini) else (sB2in);
  sB1_1=if sB1in then suB1ex(okBo,lv11,sB1in,trm,ini) else (sB1in);
  sB2=suB2en(okBo,okTop,lv11,lv9,lv13,sB2_1,trm,ini);
  sB1=suB1en(okBo,lv14,sB1_1,trm,ini);
tel

-- graph id=18 name=Top,GCTOP
node sf_2(E:event; x:int) returns(sB1,sB2,sgB,sA:bool);
var ini,lv10,lv10_1,lv12,lv12_1,lv13,lv13_1,lv8,lv8_1,lv9,lv9_1,
    okB,okB_1,okB_2,okTop,okTop_1,okTop_2,okTop_3,sA_1,sA_2,sAt,
    sB1_1,sB2_1,sgB_1,sgB_2,sgBt,trm:bool;
let
  sA_1=false -> pre sA; sgB_1=false -> pre sgB;
  sB2_1=false -> pre sB2; sB1_1=false -> pre sB1;
  okTop_1,okB_1=(false,false);
  lv10_1,lv9_1,lv12_1,lv13_1,lv8_1=(false,false,false,false,false);
  okB_2,okTop_2,lv10=
    if sB2_1 then suB2lv(E,okB_1,okTop_1,sB2_1,trm,ini)
    else (okB_1,okTop_1,lv10_1);
  okTop_3,lv9,lv12,lv13=
    if sA_1 then suAlv(E,x,okTop_2,sA_1,trm,ini)
    else (okTop_2,lv9_1,lv12_1,lv13_1);
```

```
 okTop,lv8=
    if ini and not (sA_1 or sgB_1)
    then iniu20__pointlv(okTop_3,trm,ini) else (okTop_3,lv8_1);
  sA_2 = if sA_1
         then suAex(okTop,lv9,lv12,lv13,sA_1,trm,ini) else (sA_1);
  sgB_2 = if sgB_1
         then sguBex(okB_2,lv10,sgB_1,trm,ini) else (sgB_1);
  sA=suAen(okB_2,okTop,lv8,lv10,sA_2,trm,ini);
  sgB=sguBen(okTop,lv9,lv12,lv13,sgB_2,trm,ini);
  okB,sB1,sB2=sf_4(E,okB_2,okTop,lv9,lv10,lv13,sB1_1,sB2_1,sgB,
                   sgB and trm -> (not sgB) and (pre sgB),
                   sgB -> sgB and not (pre sgB));
tel
```

**Figure 5.19:** LUSTRE code fragments for inter-level transitions

code. This has appeal since this translation would be essentially a one-to-one correspondence between semantic objects. However, the model-checking and other verification tools are unable to work with embedded C code and we lose expressive power for our system. The alternative, and harder, approach is to translate the action code into LUSTRE. The problem here is that we need to impose a sequential order on the generated LUSTRE statements which matches the execution order in the STATEFLOW. We also have efficiency problems since any values in the context not referenced by the action code have to be copied across but we are not concerned with the efficiency of the generated LUSTRE code at this point.

### Pseudo-lustre

To ease this translation we have defined a simple sequential subset of LUSTRE characterised by the following properties:

- LUSTRE statements are considered to be evaluated from top to bottom.
- Any inputs which are updated have an output value created for them[7].
- Any outputs referenced before their first definition have inputs created for them.
- Values referenced within `pre` statements are not considered as instances.
- Values on the left hand side of the equations are made unique.
- References to sequenced values on the right hand side are transformed to refer to the most recent instance.

[7] Created output variables are suffixed with the string "`_out`" (or simply "`o`" in abbreviated form) and created inputs are suffixed with "`_in`". Unique variables are suffixed by an integer.

```
-- a) Untransformed
node test(x: int)
returns(y: int);
let
  x = x + y;
  x = x + 1;
  y = y + 1;
tel
```

```
-- b) Transformed
node test(x, y_in: int)
returns(y, x_out: int);
var x_1: int;
let
  x_1 = x_in + y_in;
  x_out = x_1 + 1;
  y_out = y_in + 1;
tel
```

**Figure 5.20:** Transformation of pseudo-LUSTRE

For example, Figure 5.20 shows the transformation of a simple test node. This transformation allows us to virtually transliterate the action code directly into LUSTRE with minimal alteration. In fact, this style of LUSTRE is also useful for code generated elsewhere and is used ubiquitously in the translator.

#### STATEFLOW arrays to LUSTRE arrays

The only significant complication is arrays for which we synthesize access code which allows them to behave as variables. For example, the action code `x[0]++`, where `x` has type `int^3` is translated into:

```
xo = aset1_i(3, 0, aget1_i(3, 0, x) + 1, x);
```

We synthesize a function "`a(get|set|fill)<n>_<ts>`" for each `<n>`-dimensional array value of type `<ts>`. Note that each time an array value is accessed or updated the entire array is searched or copied resulting in very inefficient code.

#### Temporal logic operators

A similar complication arises for temporal logic code, for which we synthesize auxiliary LUSTRE routines. Figure 5.21 shows the synthesized code for the `after` temporal operator. The main issue is that the counter is only incremented when the state is active so we have to pass the associated state variable to the counter function. The code shown here simply tests if the current count (`cnt`) is greater than the required number of counts (`n`), the code for `before`, `at` and `every` being similar.

### 5.4.6 Event broadcasting

One of the most difficult aspects of STATEFLOW to translate is the generation of events *within the* STATEFLOW *model*, these are called *local* events in STATEFLOW terminology. The problem is that STATEFLOW implements these by running the interpretation algorithm to completion on

```
-- Counter function for temporal events
node sfcnts(s, E: bool) returns(a: bool; cnt: int);
var inc: int;
let
   a = s -> s and not pre s;
   inc = if a and E then 1 else 0;
   cnt = inc ->
     if a then inc
     else if s and E
          then (pre cnt) + 1
          else pre cnt;
tel


-- after function for temporal events
node sfaft(n: int; s, E: bool) returns(flg: bool);
var a: bool; cnt: int;
let
   a, cnt = sfcnts(s, E);
   flg = n >= 0 and s and (cnt >= n);
tel
```

**Figure 5.21:** Counter function for temporal logic

each transmitted local event which implies the possibility of unbounded behavior (since transmission of one event can trigger the transmission of another). On the other hand, LUSTRE provides a bounded (and known at compile time) recursion mechanism. Therefore, if we can prove (or assume) that the implicit recursion is bounded by a constant $k$, then we can translate the STATEFLOW model into a LUSTRE program with recursion bounded by $k$.

Up to now, nothing we have described implies any kind of recursive behavior in the translator, we could simply generate the code by preserving the hierarchy in the original STATEFLOW model. Now, however, we have to know the arguments to the top-level call when we implement a broadcast event. We could either make the translator a fix-point computation where the arguments to previously generated graphs are updated when event broadcasts happen, we could use a two-pass method where the first pass computes the nodes and arguments and the second generates the code or, since the *only* recursion point is the top-level call we could simply predict the arguments to this node and use that for the broadcasts. Currently, we use the last (and simplest) option but if, for example, we were to implement the `send` function, which is the ability to send an even to a specific state, as a function call to the relevant node we would require a more general analysis.

Another slight complication is that LUSTRE will not accept a constant value for the top-level node. Bounded recursion requires the presence of a recursion variables which have to be statically evaluated. We thus generate a proxy node for the top-level call and seed this with the value of the recursion variable. We implement bounded recursion by creating a `const`[8] recursion variable for event broadcasts which we call the "event stack size". We can then call the top-level node at the point where an event is broadcast, reducing this constant by one. This allows emulation of the recursive nature of STATEFLOW's interpretation algorithm *up to a finite limit set by the event stack size*. If we have a proof of the bound on event broadcast recursion then our behavior will be the same as STATEFLOW's.

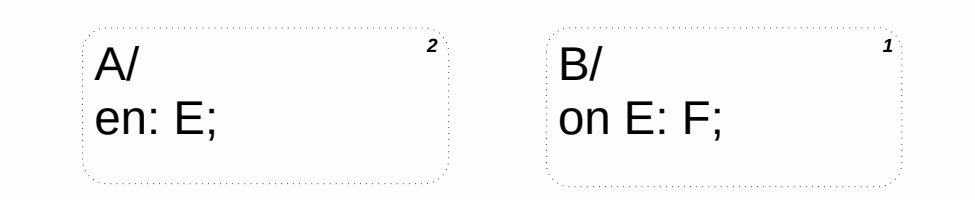



**Figure 5.22:** A model with non-confluent parallel states requiring event broadcasting

In Figure 5.22 the two states `A` and `B` are evaluated in the order `B` then `A` but `A` emits event `E` whereas `B` receives it. Figure 5.23 shows the relevant parts of the generated code. The event broadcast routines simply call the recursion point (`sf_2ca`). At the point of call, all events are cleared (`clr`) and the event being broadcast is set (`set`). The recursion point is the `sf_2ca` node and the top-level function (`sf_2`) is simply a wrapper for `sf_2ca` replacing the recursion

[8] A `const` value in LUSTRE is not actually a constant. It refers to a value which can be statically evaluated at compilation time.

variable (`const n`) with the event stack size. This is needed because LUSTRE will not accept a `const` value as an input to the top-level node.

```
-- entry action for node id=3 name=A
node enaA(F,E: event; sB,sA,term,init: bool;
          const n: int)
returns(Fo: event; sBo,sAo: bool; Eo: event);
let
  Fo,sBo,sAo,Eo=
    with n=0 then (F,sB,sA,E)
    else sf_2ca(clr,set,sB,sA,term,init,n-1);
tel


-- graph id=7 name=Parallel5,call
node sf_2ca(F,E: event; sB,sA,term,init: bool;
            const n: int)
returns(Fo: event; sBo,sAo: bool; Eo: event);
...


-- graph id=7 name=Parallel5,top
node sf_2(dummy_input: bool) returns(F: event);
let
  ...
  F,sB,sA,E=sf_2ca(F_1,E_1,sB_1,sA_1,term,init,1);
tel
```

**Figure 5.23:** Code showing event stack

Within this scheme it is possible to implement STATEFLOW's "early return logic" which is intended to reduce the possibility of inconsistent states arising from the misuse of event broadcasts. It results, however, in messy and inefficient code since virtually all activity after the potential processing of an event has to be guarded with a check of the parent or source state. This has been partially implemented in our translator, for example, in the above code, if state `A` was within another state, say `A1`, then the call to the entry action for state `A` would actually be something like:

```
if (sgA1 and enA) then enaA1(...);
```

This static recursion technique allows us, in theory, to emulate the behaviour of STATEFLOW charts which exhibit bounded-stack behaviour. In practice, there is a heavy penalty to pay for static recursion since the recursion encompasses practically the entire program. This means that each event broadcast point results in expansion of the whole program at that point, down to the level of the event stack. Practical experience with the translator shows that an event stack size of 4 is about the greatest that can be accomodated in reasonable space and time.

Finally, we can easily accommodate STATEFLOW's `send` facility which allows sending of an event to a named state. One possibility is to view this as a function call of the target state [HR04], however, this would require generalization of the recursion mechanism to allow calls to intermediate nodes. A simpler solution is to simply treat events as integers and use the convention that 0 is an inactive event, 1 is a broadcast event and events with other integer values are targeted at the state with that identity number. For this purpose we abstract the `event` type and provide constants for event testing:

```
type event = int;
const set = 1; clr = 0;
```

The `on` action for state `B` (id number 4) would thus become guarded by:

```
if ((E = set) or (E = 4)) then ...
```

### 5.4.7 History junctions

History junctions are a STATEFLOW feature which allow states to "remember" their previous configuration in between activations. This is easily handled by our translator by keeping local variables within each node corresponding to a state with a history junction. The only complication is how to trigger storage and restoring of the history values. Luckily, the `init` and `term` flags correspond almost exactly to the semantics of history junctions, we only need to store them when `term` is `true` and restore them when `init` is `true`. Figure 5.24 shows the relevant code.

```
node sf_3(sAin,sBin,sgTOP,term,init: bool)
returns(sA,sB: bool);
var sAh,sBh,...: bool;
let
  sB_1,sA_1=(false,false) ->
    if init then (pre sBh,pre sAh) else (sBin,sAin);
  ...
  sBh,sAh=(false,false) ->
    if term then (sBin,sAin) else (pre sBh,pre sAh);
tel
```

**Figure 5.24:** Saving and restoring history values

### 5.4.8 Translation fidelity

It is not possible to formally verify the equivalence of STATEFLOW's and our translator's behaviors, principally because of a lack of a formal definition for STATEFLOW. Our translator was developed, however, directly from the STATEFLOW documentation and its description of the interpretation algorithm which, as far as possible, we have encoded into LUSTRE. We have also

manually verified the equivalence of the two systems on a substantial set of example STATEFLOW models based around the subset of STATEFLOW which we currently support. From our point of view, however, the primary reference for the behavior of the translated code is the LUSTRE translation. In a real-world example we would perform tests and validation upon the LUSTRE code and not upon the STATEFLOW model directly.

This also applies to our link with SAFE STATE MACHINES (SSM) by *Esterel Technologies, Inc.*. SSM, like STATEFLOW, is also a graphical interface to a finite state machine system but, unlike STATEFLOW, is based on a sound formal semantics and there exists a formal translation path into languages such as LUSTRE. The question exists, however, as to how to translate legacy STATEFLOW code into SSM and the issues embodied in our translator also apply to translation from STATEFLOW into SSM. Our translator can, however, be used as a reference semantics for this translation since its output should, in theory, have the same semantics as the output from STATEFLOW → SSM → LUSTRE.

## 5.5 Which subset of STATEFLOW do we translate

Currently, we can translate hierarchical and parallel AND states assuming *no* inter-level transitions. We can implement event broadcasting provided the broadcasting recursion is bounded by a reasonably small value. State entry, exit, during and on-actions as well as condition and transition actions for transitions are all supported. Only part of the action language is translatable but we can implement array processing and so-called *temporal logic operators*. This gives basic functionality. In addition, however, we can implement sending of events to specific states, history junctions and inner transitions.

## 5.6 Enlarging the "safe" subset by model-checking

The existence of a translation from STATEFLOW into LUSTRE allows us to immediately apply the existing model-checking tools for LUSTRE to STATEFLOW models.

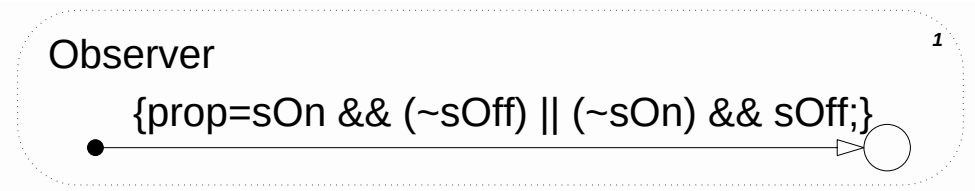



**Figure 5.25:** Simple observer in STATEFLOW

For example, Figure 5.25 shows a simple observe implemented in STATEFLOW for the model in Figure 5.6. Here the property is a trivial mutual exclusion of states and LESAR verifies this property without consuming any significant time or memory.

In this section we demonstrate two useful properties that can be model-checked in STATEFLOW models, i.e., confluence of parallel states and boundedness of event broadcasting. Our translator is able to generate auxiliary LUSTRE nodes which are observers for properties supplied to the translator. Currently, these are LUSTRE expressions but it should be possible to allow these expressions to be supplied by the STATEFLOW model in the form of graphical functions or some other form of annotation. This would obviate the necessity of the user learning LUSTRE's syntax and semantics. In this section we simply demonstrate two useful properties that can be model-checked in STATEFLOW models, i.e., confluence of parallel states and boundedness of event broadcasting.

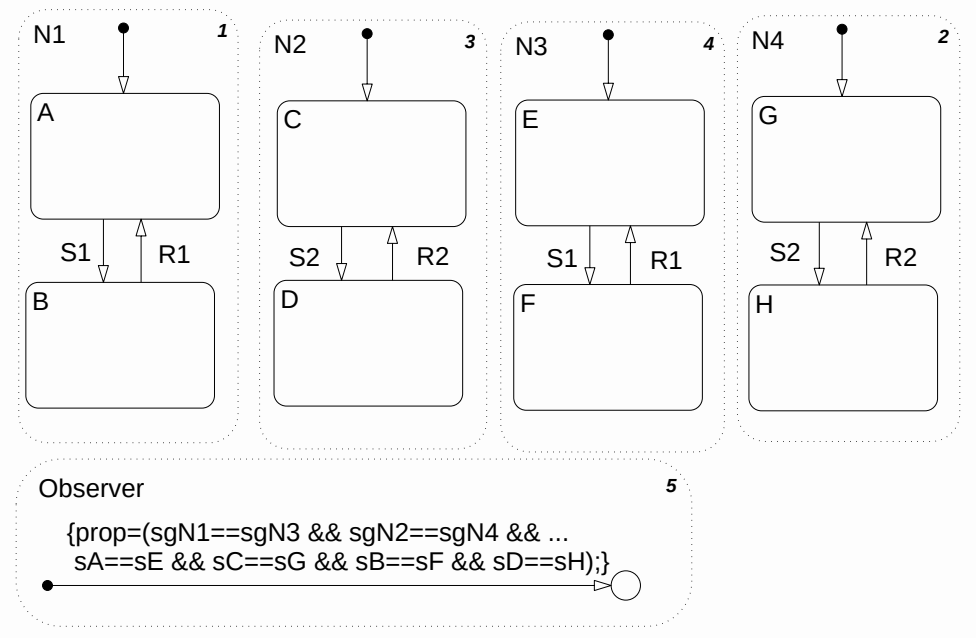



**Figure 5.26:** An observer for parallel state confluence

Figure 5.26 shows a set of parallel states[9]. States `N1` and `N2` (executed in the order `N1` then `N2`) and states `N3` and `N4` (executed `N4` then `N3`) form two versions of the same simple machine except for the order of parallel execution. The figure also shows an observer which directly compares equivalent state variables between the two machines. Running LESAR on the generated LUSTRE code results in a `TRUE` value so we can deduce that the order of execution of parallel states in the machine `N1`/`N2` (or `N3`/`N4`) is irrelevant.

Figure 5.27 shows a STATEFLOW chart which requires either parallel state confluence or the use of an event stack. State `TOP1` generates a local event `E` upon receiving input event `G`. Event `E` is received by state `TOP2` which then emits output event `F`. To allow detection of event stack overflow the translator generates an additional local value "`error`" which is set if there is an

[9]The state variable names are accessible in our translator so `sOn` refers to the variable for the `On` state. These *pseudo-variables* have to be included in STATEFLOW's data dictionary.

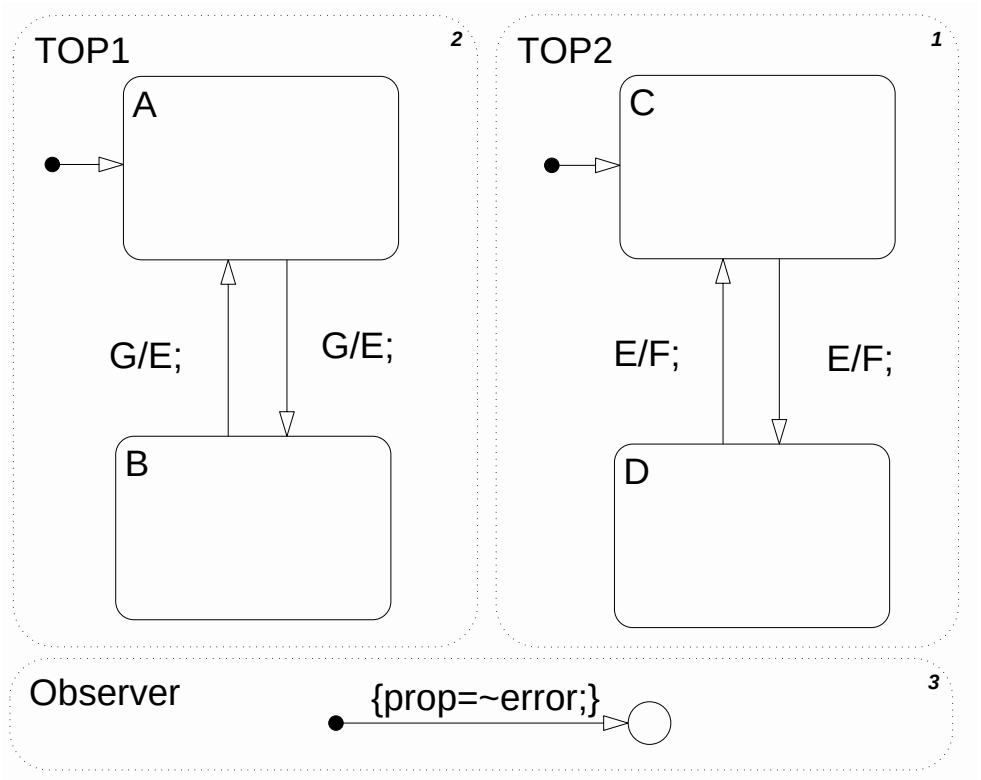



**Figure 5.27:** An observer for event stack overflow

attempt to broadcast an event when the event stack counter is zero. The broadcast statement for event F is show in Figure 5.28.

```
propo,Fo,sAo,sBo,sCo,sDo,sgObservero,sgTOP1o,sgTOP2o,
erroro,Eo =
  with n = 0
  then (prop,F,sA,sB,sC,sD,sgObserver,
        sgTOP1,sgTOP2,true,E)
  else sf_2ca(clr,clr,set,prop,sA,sB,sC,sD,
              sgObserver,sgTOP1,sgTOP2,error,
              term,init,n-1);
```

**Figure 5.28:** Code for event broadcast with error detection

If TOP2 is executed before TOP1 we need event broadcasts to allow E to be received by TOP2. Furthermore, if output event F is to be broadcast we need a minimum event stack of 2 which is verified by LESAR. Model-checking using the error property gives a FALSE property for an event stack depth of 1 but a TRUE property if the event stack is set to 2. Finally, if we reverse the order of execution of states TOP1 and TOP2 we can get a TRUE property with an event stack size of zero.

Although these examples are trivial the analysis itself can be extended to models of any complexity. We envisage using the model-checking not just for verification of safety properties but also as a means of enhancing the subset of STATEFLOW which we are able to implement. A designer can use model-checking to spot where his design does not conform and where to fix the model to bring it into conformance.

## 5.7 Related Work

STATECHARTS [Har87] are sometimes compared with STATEFLOW since both are visual representations of state machines. STATECHARTS semantics are complex to define, in fact, many variants exists, for instance, see [HN96, HG97] and references therein. There has also been much work into either translating STATECHARTS into a known system such as hierarchical automata [MLS97] or by deriving a semantics for a suitable subset [LvdBC00]. Most of these works yield semantics that are different from those of STATEFLOW. For example, STATEFLOW has no notion of true concurrency since its semantics is "run-to-completion".

UML's *state-machine diagrams* also resemble both STATEFLOW and STATECHARTS. Notice that UML state machines have a "run-to-completion" semantics, like STATEFLOW [Gro05, HG97].

Tiwari [Tiw02] describes analysis for SIMULINK/STATEFLOW models by translating them into communicating pushdown automata. These automata are represented in SAL [BGL+00] which allows formal methods such as model-checking and theorem proving techniques to be

applied to these models. Essentially, the system is treated as a special hybrid automaton and algebraic loops involving STATEFLOW charts are not considered.

[Nee01] presents the data model of SIMULINK and STATEFLOW as a UML class diagram.

Hamon and Rushby have developed a structural operational semantics for STATEFLOW [HR04] for which they have an interpreter to allow comparison with STATEFLOW. Their subset of STATEFLOW seems to have been inspired by the Ford guidelines [For99], for instance loops are forbidden in event broadcasting and local events can only be sent to parallel states. They have other restrictions as well, such as forbidding transitions out of parallel states but in general support most of the STATEFLOW definition including inter-level transitions. They also have a translator into the SAL system which allows various model-checking techniques to be applied to STATEFLOW.

## 5.8 Conclusions

In this Chapter we studied the translation of STATEFLOW to LUSTRE. Although the behavior of STATEFLOW is deterministic, it may lead to problematic designs, for instance, such that a simulation cycle does not terminate or where tiny changes in the graphical layout may modify the behavior of the model. To handle these issues we proposed a set of conditions, that can be checked statically, that guarantee that a STATEFLOW model will be free of the semantical problems described.

Besides the analysis of STATEFLOW and those static rules, we have provided a method to translate a STATEFLOW Chart into the synchronous programming language LUSTRE. This translation can be used in many different ways. First of all, for code generation, using one of the code generators available for LUSTRE. The translation can also be used for verification of STATEFLOW models, using LUSTRE's model-checker Lesar. We can use verification to check not only general properties about the model but also specific properties, for instance, those related to semantical issues above (e.g., check the confluence of a set of states).

The translation algorithms presented in this and the previous Chapter have been implemented in a prototype tool called SF2LUS. This is presented in the Chapter that follows.

# Chapter 6

# The SIMULINK/STATEFLOW to LUSTRE Translator: Tool and Case Studies

We implemented the algorithms discussed in Chapter 4 and Chapter 5, in prototype tools that translate SIMULINK and STATEFLOW into LUSTRE. Those tools are the S2L and SF2LUS respectively, and on top of them we added a coordination system able to translate a model that contains both SIMULINK and STATEFLOW parts. This latter, mostly scripting, tool is called SS2LUS as for *Simulink-Stateflow-to-Lustre*.

The tools can be downloaded free of charge, after signing an academic license[1]. To date, the tool has been distributed to five different users and we have a collaboration with most of them in terms of bugs and new features needed.

In this Chapter we study certain aspects of the implementation of those tools and the way they interact. Moreover, we demonstrate their use on a number of real models we tested during implementation phase.

Note that in Appendix A, we provide the output of the tools when invoked using the `--help` flag. This output contains the main information about the tools and all the possible invocation arguments.

## 6.1 Tool

### 6.1.1 The S2L tool

In Chapter 4, we studied all the aspects of the translation from SIMULINK to LUSTRE. Alongside to studying those methods and algorithms we implemented them in a prototype tool that achieves the goals of this translation, modulo all the constraints that are discussed in Section 4.3.

This tool is named S2L as for *Simulink-to-Lustre* translation and it is a tool written using the Java programming language. Though the initial implementation was back and forth compatible with older Java versions, in the latest updates of the tool, we used the latest addition of Java, the

[1] For more information refer to the official website of the tool following the *Simulink, Stateflow* link in the following web page: http://www-verimag.imag.fr/SYNCHRONE

safety of the software. This latest addition in S2L will require, to compile and run, a version of Java greater or equal to `Java 1.5`.

All the above packages are documented using the Java's *JavaDoc* documentation tool and its appropriate notation. In fact we provide two flavors of *JavaDoc* documentation, the one is for private development usage, with all the details of private classes and a "light" version, where we only describe the public classes that one will use to build on top of this ADT some new behavior.

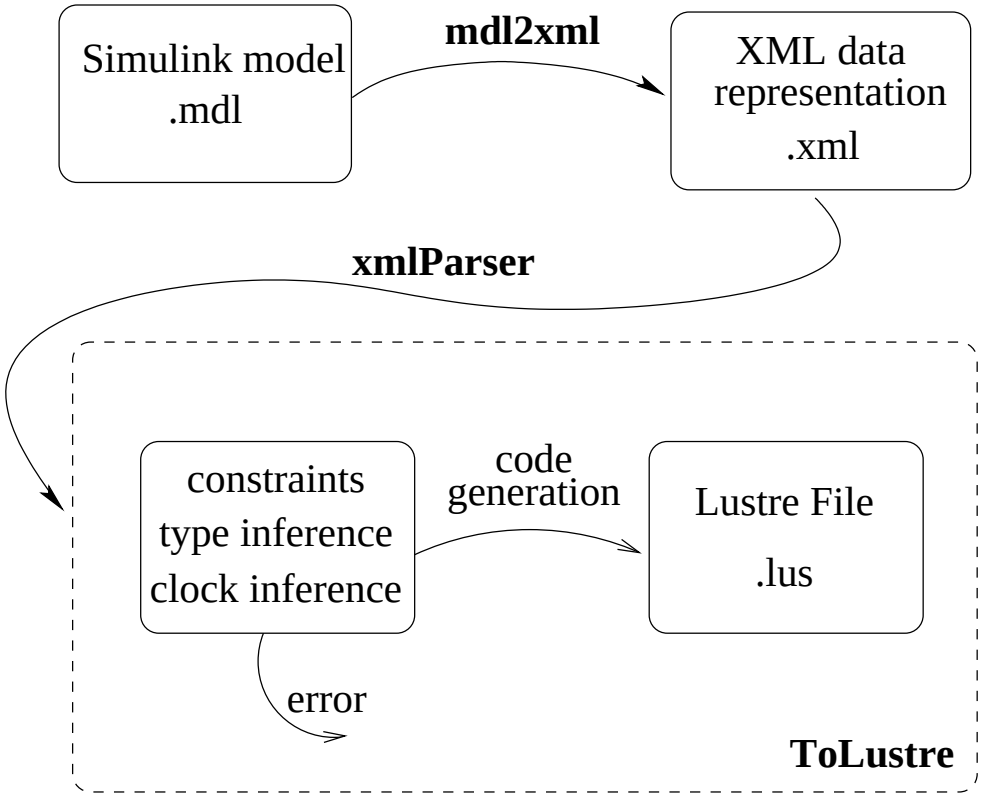



**Figure 6.1:** The S2L tool architecture.

In Figure 6.1 we can see the software architecture of the S2L. It is composed out of three main packages:

**fr.verimag.mdl2xml** This package makes a simple one-to-one translation from the SIMULINK model file to an equivalent *XML* representation. However, this package can be invoked with a parameter that will filter the irrelevant to the translation information such as, dimensions and placement of blocks in the SIMULINK graph.

In Appendix B, we provide an example of this translation to *XML*.

**fr.verimag.xmlParser** This is the package that will parse the *XML* file and generate the Abstract Data Tree holding all the information of the model in its structures. All the information

is stored with in the ADT with root node an instance of the `class Model`. Other, main classes–components of this tree are `Block, Line, SimulinkSystem, Port`.

**fr.verimag.s2l** In this package we have implemented all the inferences and the code generation algorithms that we have seen in Chapter 4. The main class is the `class ToLustre`, which instantiates and calls the appropriate classes and methods, found on that package.

During the implementation of the tool, *MathWorks* has released a number of new releases of the MATLAB toolkit, changing the model file format from one release to the next. This made it difficult for the tool to follow all the changes, so we decided to stick with the file format introduced by MATLAB-6 (releases r12, r12.1 and r13)[2].

The current version of the tool is a command line program. However there is a graphical user interface version, which currently is in a Beta and not stable version. The GUI is also implemented in Java and the main window you can see in Figure 6.2. The steps for the compilation of a SIMULINK model are (1) first we chose the file to translate, (2) we invoke the translator pressing the "S2L" button (the user will be asked to give the translation flags and options then). Moreover, (3) we can continue (if we have a LUSTRE complier) by passing the resulted LUSTRE generated code to that complier.

As we see in Figure 6.2 in the upper, gray text-box we have the resulted LUSTRE code and in the bottom, yellowish text box we have the output of the translator and the LUSTRE compiler.

Thus, for the command line S2L the input arguments must imperatively contain the model (.mdl) file and also a number of optional flags and options that alternate certain features of the translation. Below we summarize the most important of them (for a complete list of the arguments the reader may refer to the Appendix A), as this is provided by the tool's help message:

**–reluc** Using this flag will generate LUSTRE code that respects the syntax of the RELUC compiler. RELUC [3] is an experimental LUSTRE compiler provided by ESTEREL. The default LUSTRE output format is the one that respects LUSTRE V4 distribution.

**–debug** This option will generate on the standard output debug information of the translation procedure. Moreover, a file named `allVars`, will be generated on the same directory containing hierarchically all the LUSTRE variables generated along with their timing and type information as they have been inferred from the inference algorithms that we saw in Section 4.4 and Section 4.5.

**–monoperiodic** Mainly for debugging purposes we may need to bypass the clock inference, in which case we use this option which ignores the Sample Times of all the elements of the SIMULINK model, or better, treats everything as being inherited (or $-1$). This means that the triggered subsystems will still have the correct triggering clock as their basic clock.

[2] Note that some times, using a newer version of MATLAB and trying to save in an old format, will introduce elements in the model file that are not supported by the respective version itself and thus by our tool. So we strongly propose the use of MATLAB 6 - release13 to obtain the best results.

[3] RELUC stands for Retargetable Lustre Compiler

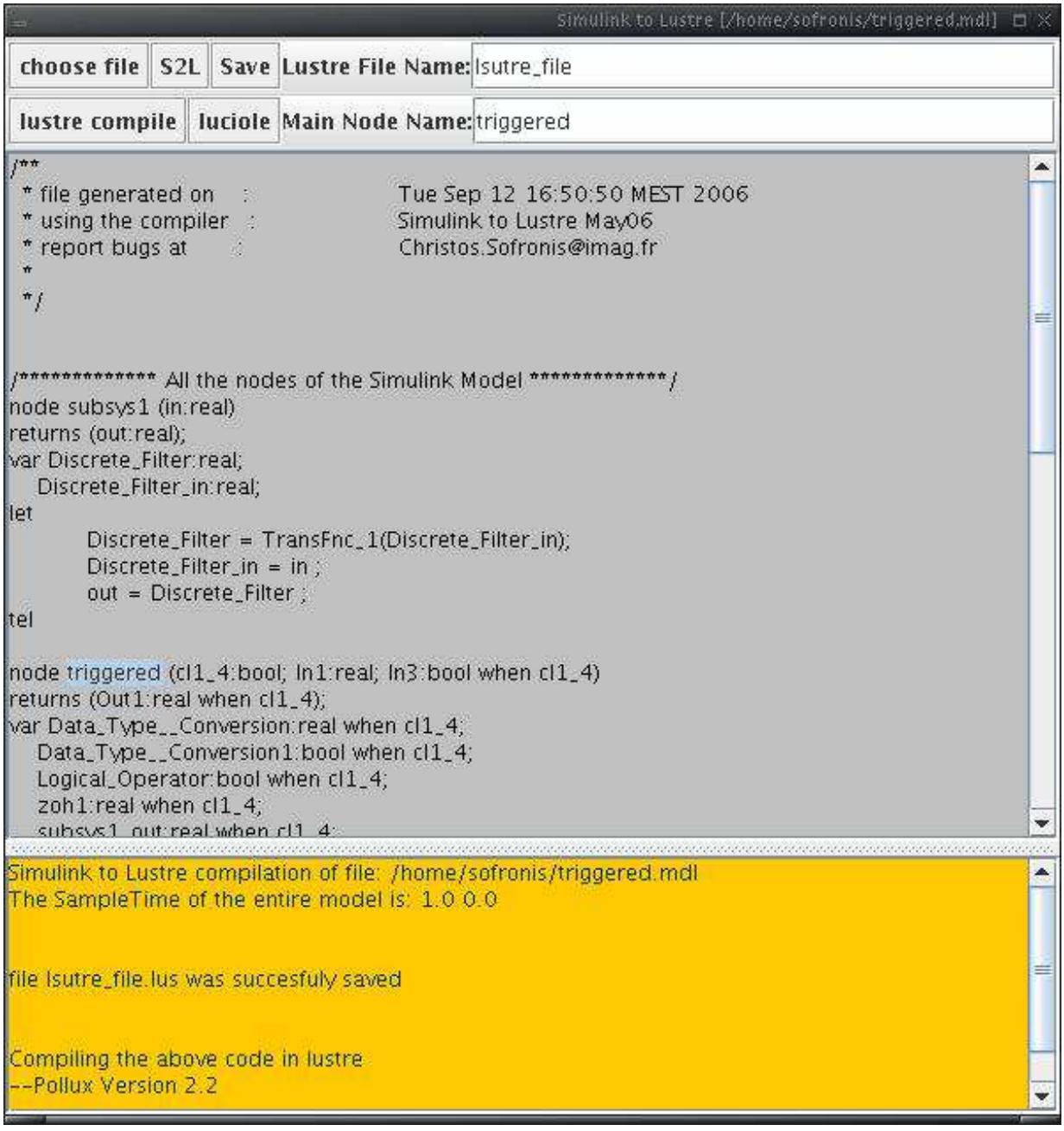



**Figure 6.2:** The main window of the GUI version of the S2L tool.

**–choose-system** In a big and complex system, we may want to translate only a small part hidden in the tree structure of subsystems. Using this flag we interactively choose the subsystem that will be the root node of the translation to LUSTRE.

**–no-main** By default, S2L, will generate a folding node that encapsulates the main node in the generated LUSTRE file. This helps on generating internal clocks and associating them to input/output LUSTRE variables and thus giving a "cleaner" interface to the user that no longer he has to cope with timing relation between inputs. However, this may be redundant if there is no timing variance between inputs and/or outputs, thus using this option will skip this step.

**–xml** Using this option, the tool only generates the *XML* representation of the SIMULINK model and stops the execution.

The result of invoking S2L is to generate a LUSTRE program in a `.lus` file (usually with the same file name, unless explicitly stated). Moreover, the tool outputs the period that the LUSTRE program should be run at, so that the translation corresponds to the source model. As we have seen in Section 4.5, this is a result of the clock inference algorithm.

However, if during any stage of the translation there is an error, then the translation is aborted and an error message is printed on the standard output. Note also, that the code generation is done *on-the-fly*, which means that every node that is generated is written and flushed to the output file, so that in any undesired exception, we will still have the LUSTRE code generated up to the occurrence of this exception.

### 6.1.2 The SF2LUS tool

The translation of STATEFLOW to LUSTRE studied in Chapter 5 has been implemented in the SF2LUS tool. This tool has been implemented by Norman Scaife, a post-doc at Verimag from 2003 to 2005. SF2LUS is written in *Objective Caml* (OCaml).

SF2LUS does not use intermediate *XML* representation and the ADT is parsed immediately out of the model file. On the other hand SF2LUS provides more options to toggle features of the translation and produce detailed debugging output. Note that the tool initially was conceived to parse the STATEFLOW part of a model and construct a graphical representation of it. Then various methods for generating code were implemented and the final version, actually implements all the features we have studied in the previous chapter.

The input to the tool is a STATEFLOW model. The tool supports the MATLAB 7 release 14 and if the file format remains the same in the next versions of the tool, it will support them also.

Output from the tool is primarily LUSTRE V4. This version of LUSTRE includes bounded iteration using the `when` construct in conjunction with statically-evaluated constants and proved very useful in structuring STATEFLOW's unbounded iterations such as event broadcasts and junction loops. In general, however, it is intended that SF2LUS be used in conjunction with the planned STATEFLOW analysis tool to allow the elimination of unbounded recursion either by automated transformation or by manual editing of the STATEFLOW chart.

Other outputs include preliminary versions of SCADE [Est] which allows STATEFLOW charts to be included as source code into the SCADE suite of tools by *Esterel Technologies, Inc.*, RELUC which is a commercial version of LUSTRE also by ESTEREL and some minor support required by the abstract interpretation tool NBAC (via LUS2NBAC provided in the LUSTRE distribution). These output formats are only partially supported, however. Note that the LUSTRE output itself can be used as input to a variety of tools including the model-checker LESAR [RHR91, HLR92].

Hereafter we discuss the principle options of the translator, invoked in command line as arguments.

### Output language

Four output languages are currently supported, triggered by the following options:

**-pollux** Generate LUSTRE V4 output. This is the default mode and the most fully supported.

**-nbac** Generate NBAC output. NBAC is almost completely compatible with LUSTRE V4 via the LUS2NBAC utility supplied with the LUSTRE distribution. This option simply triggers emulation of the integer modulus function which seems to be missing from NBAC.

**-reluc** Use RELUC modifications. Again, since the SF2LUS translator uses only a small subset of LUSTRE the output is almost compatible with RELUC. Currently, this option triggers some additional parenthesization which seems to be needed. Currently, arrays and the event stack are not supported since RELUC does not have LUSTRE's static recursion mechanism.

**-scade** Use SCADE modifications. There are some syntactic differences between SCADE and LUSTRE, some of which can be ironed out by a simple transformation on the output. These involve constructs such as:

```
(x,y) = if p then f(a,b) else (c,d);
```

which are not supported by SCADE. Again, there are no arrays or event stack.

### Namespace management

Namespace management in the SF2LUS translator is not fixed, for several reasons:

- There is the tension between providing human-readable LUSTRE output without making the code too verbose.
- Different users may have different preferences as to what is readable, depending upon their intended usage of the resulting code.
- It is difficult to translate namespaces accurately between two such widely differing languages as STATEFLOW and LUSTRE. It is simple to convert names from one syntax space to another but doing so while retaining the flavor of the original language is difficult.

- It is possible that different translations may have to coexist in the same context which will inevitably result in namespace collisions. This is exacerbated by the fact that LUSTRE V4 has no concept of modularity.

For these reasons, SF2LUS supports several options controlling the way the output namespace is generated:

**-names** Use state names in variables. So, for example, a state called `A` generates a variable `sA` (or `state_A`).

**-ids** Use state ids in variables. A state with `id` 3 will be referenced by `s3` (`state_3`).

**-names_ids** Use both names and state ids in variables. For example `s3_A` (`state_3_A`).

**-long_names** Use unabbreviated names in the output. Currently, the complete list of abbreviations is:

| | | | | | | | |
|---|---|---|---|---|---|---|---|
| _point | | action | a | after | aft | at | at |
| before | bfr | call | ca | change | ch | condition | c |
| count | cnt | counter | cntr | counts | cnts | during | du |
| end | end | enter | ent | entry | en | error | err |
| event | ev | events | evs | every | evry | exit | ex |
| flag | flg | graph | g | history | h | in | in |
| increment | inc | init | ini | inners | ins | junction | j |
| link | l | okay | ok | out | o | pre | p |
| print | pr | property | prop | state | s | stateflow | sf |
| stub | st | subgraph | sg | term | trm | tmp | t |
| transition | tr | update | u | valid | v | verify | verif |

**-varprefix** Prefix all variables (for namespace conflict avoidance). Do not use this, it is present for debug purposes only.

**-prefix,-suffix** Prefix/suffix all toplevel names. This is used, for instance, when one wishes to compare the output from two different translations. All the visible identifiers in the output code are prefixed by the given string, for example, "`-prefix A`" might give:

```
type Aevent = bool;
const Aset = true; Aclr = false;
node Asf_2(Set, Reset: Aevent) returns(sOff, sOn: bool);
```

### General translator control

These options control the translation process at the most basic level.

**-no_self_init** Top level graph does not provide initialization. Normally, the `init` and `term` flags are automatically set to:

```
init = true -> false;
term = false;
```

at the top level of the code. This option disables this behavior but is only present for debugging the translator.

**-ess** Event stack size. This sets the depth of the event stack, see Section 5.4.6 for usage information and *caveats*.

**-sends** Enable sends to specific states. This option triggers some additional processing which allows STATEFLOW's `send` function to be implemented. Note that events become integers which may affect subsequent analysis and that currently this feature is only partially implemented. See Section 5.4.6.

**-junc_states** Treat junctions as states. When this is set junctions are given a physical state (called `j<ID>` or `junction_<ID>`) and the chart can stay in a junction after a reaction. An additional output boolean (`v` or `valid`) is generated which is `true` if and only if the current state is not a junction. See Section 5.4.2.

**-errstate** Add error processing to event broadcasts. This generates an extra output boolean variable (`err` or `error`) which is set to `true` if an event is broadcast at the lowest level of the event stack. This logic is switched off if the event stack size is zero.

**-unroll** Unroll loops according to loop counters. Using the annotations in the STATEFLOW chart described in Section 5.4.2 transition networks involving loops are unfolded a fixed number of times resulting in a loop-free chart. The unrolling algorithm is currently very primitive and has complexity problems.

### Data management

Handling MATLAB's workspace is complicated by the fact that it is stored in a binary format which external tools cannot read. Hence, the translator has to make some assumptions about the workspace which it communicates via the `.mws` file. Note also that this file provides the means of communication with the S2L tool also. These options allow some additional control over workspace values:

**-create_missing** Add missing data to data dictionary. If a chart contains a reference to a variable not in STATEFLOW's data dictionary then it can be automatically created. All such variables have to have the same scope and type, however.

**-missing_scope** Scope for missing data (default: `INPUT_DATA`). Recognized values are:

| | | | |
|---|---|---|---|
| r13/r14 | `INPUT_EVENT` | `OUTPUT_EVENT` | `LOCAL_EVENT` |
| | `OUTPUT_DATA` | `INPUT_DATA` | `LOCAL_DATA` |
| | `TEMPORARY_DATA` | `CONSTANT_DATA` | |
| r14 | `FUNCTION_INPUT_DATA` | `FUNCTION_OUTPUT_DATA` | `PARAMETER_DATA` |

**-missing_datatype** Data type for missing data (default: `double`). Known values are (not all are supported):

```
double   single  int8    int16
int32    uint8   uint16  uint32
boolean  fixpt   ml
```

**-no_constants** Omit workspace constants from output. This is used by SS2LUS and prevents SF2LUS from including constants defined in the MATLAB workspace file from being included. The output is not legal LUSTRE since the constants are expected to be provided by S2L.

### Time

The STATEFLOW implicit time variable `t` is slightly problematical since LUSTRE does not have any notion of absolute time. For stand-alone STATEFLOW-generated LUSTRE code the following options allow `t` to be generated automatically assuming a fixed time difference between reactions. If time is not emulated here then it is assumed to be an input to the chart. The MATLAB workspace file contains an entry indicating whether `t` is an input or not.

**-emulate_time** Provide internal time value. References to STATEFLOW's time value are implemented internally in the LUSTRE code according to the following two options.

**-start_time** Start time for emulated time (default: 0.0). The value of `t` at reaction zero.

**-time_increment** Time increment for emulated time (default: 1.0). The `t` variable is incremented by this amount at the start of each reaction.

### Observers

One of the main uses of the SF2LUS translator is in the proof of safety properties. The following options support this activity:

**-observe** Add observer node for given expression. Generate a single LUSTRE node observing the expression given as a string of LUSTRE code. State variables can be observed provided the `-states_visible` option is set.

**-no_observers** Don't read observer file. By default, SF2LUS looks for a file `<file>.obs` when given a model file `<file>.mdl`. If it exists it is assumed to be a file containing observable expressions, one per line. LUSTRE-style comments (`--`) are permitted.

**-consistency** Add a state consistency observer. This option causes an observer for the state variables to be generated. The observer is called `consistency_<CID>` and is mostly used to verify the translation process, see Section 5.6. The `-states_visible` option is set automatically.

**-counters** Add loop counters to junction networks. An additional integer output for each junction in the chart is generated. The counters are called `cntrj<ID>` where `<ID>` is the `id` number for the junction. Each counter is incremented when its junction is entered during transition path analysis. Currently, maximum values are not maintained so the values have to be checked after each reaction. In addition, an observer (called `loop_counters_<CID>`) is generated for all the junctions annotated as in Section 5.4.2. The `-junc_states` option is set automatically.

**Debugging features**

Finally there is a number of options used to help in debugging the translator. Some may be of use in debugging STATEFLOW charts, however. Given the large number and the specialization of those options, we recall the reader to refer to the tool usage itself for further information.

### 6.1.3 SS2LUS tool architecture and usage

In this Section we will present the SS2LUS tool, which translates models composed by both a SIMULINK and a STATEFLOW part. SS2LUS, written in the *shell*'s scripting language, invokes the two tools S2L and SF2LUS and interface them in a "clean" manner: whenever S2L finds a STATEFLOW block, it submits it to SF2LUS which translates it into a LUSTRE node and returns this node (body plus type signature) back to S2L. Thus, SS2LUS is a mechanism that combine the functionality of those two independent tools and accepts he following inputs:

- The SIMULINK/STATEFLOW model to be translated to LUSTRE. This is a `.mdl` file and it must be generated by a MATLAB release between $r12$ and $r14$.
- Arguments to be passed to S2L.

The output is a LUSTRE program (a `.lus` file) and the period that the LUSTRE program must be executed, which as we have seen in Section 4.5, it is the output of the clock inference algorithm. Also the above discussed `.mws` file is generated in the current directory.

The procedure of the translation is to call the module to translate the STATEFLOW part and generate the LUSTRE code in a temporary file. Upon correct execution and translation of the STATEFLOW part[4] there is the translation of the SIMULINK part. However in that case, there is a primitive exploration of the STATEFLOW part, from S2L to extract the inputs and outputs of the STATEFLOW chart, so that the correct node call is generated.

Upon successful completion of the SIMULINK translation also, we append the LUSTRE code of the temporary file and we have the LUSTRE program that corresponds to the initial model.

Though the independent tools have several parameters and flags to alternate features of the translations, the global tool have restricted this usage to the default, which is the intersection so that both tools can cooperate. However, one may use the tools independently and generate the glue code between them manually, i.e., the code that will call the STATEFLOW part from within the SIMULINK part.

[4] In any case that an error occurs, an error message is printed and translation is aborted

## 6.2 Case Studies

In this section we present four case studies of the use of the above tools. We have used S2L to translate two SIMULINK models provided by the Audi automotive constructor in the context of the European IST project "NEXT TTA".[5] The first model defines a warning processing system and is part of a larger controller used by Audi in production vehicles. The second model is a larger steer-by-wire controller. Both models have been given to Verimag for internal project purposes and cannot be made public due to intellectual property restrictions. Thus, we only present parts of these models.

In Section 6.2.3, we present a case study that we have created to test the SF2LUS translator and in Section 6.2.4 there is an case study of our overall translator, which generates one LUSTRE program out of a model composed by both a SIMULINK and a STATEFLOW part.

### 6.2.1 Warning processing system

The objective of the system is to recognize if a car is moving towards its physical limits and to generate a warning to the driver. However, the warning may be canceled because of a number of conditions ensuring that the car is not in a dangerous situation.

The main subsystem is called `cancel_warning` and is shown in Figure 6.3. The subsystem has ten inputs and five outputs. Inputs `warning_1_in` and `warning_2_in` are generated on the same *electronic control unit* (ECU) and are sampled with a rate of 20 milliseconds (ms). Each warning is a boolean, representing an a-priori need to issue a warning. The input signals `warning_1_amp_in` and `warning_2_amp_in` are the amplitudes of the warnings. They are 8-bit signals sampled with a rate of 20 ms. The rest of the input signals (`signal_1`, `signal_2`, ...) are used to compute the warning permission `no_warning` and they have a sampling rate of 4 ms. The outputs of the main subsystem are the warning signals and their amplitudes (with the sampling rate of 20 ms) as well as the warning permission (with the sampling rate of 4 ms).

The `cancel_warning` subsystem is itself composed of a number of subsystems and basic blocks, as shown in Figure 6.4. The subsystem `no_warning_conditions` is supposed to check a number of conditions which cancel the warning. Depending on its output, the subsystem `eliminate_warning` filters the warning. The subsystem `warning_filter` blocks one of the warning signals if the other one is already active. The subsystem `warning_duration` ensures that a warning which has been issued will be sustained long enough in order to be perceived by the user.

The entire SIMULINK model has a hierarchy depth of 6 layers (including the top-level system). It contains 20 subsystems and more than 200 total components (including subsystems, basic SIMULINK blocks, as well as input/output port blocks).

Translating the entire SIMULINK model to LUSTRE using the S2L tool takes less than a second. The resulting LUSTRE program is 718 lines long. A small part of the program is shown in Figure 6.5. It contains the signature of the LUSTRE node corresponding to the subsystem

[5] http://www.vmars.tuwien.ac.at/projects/nexttta/

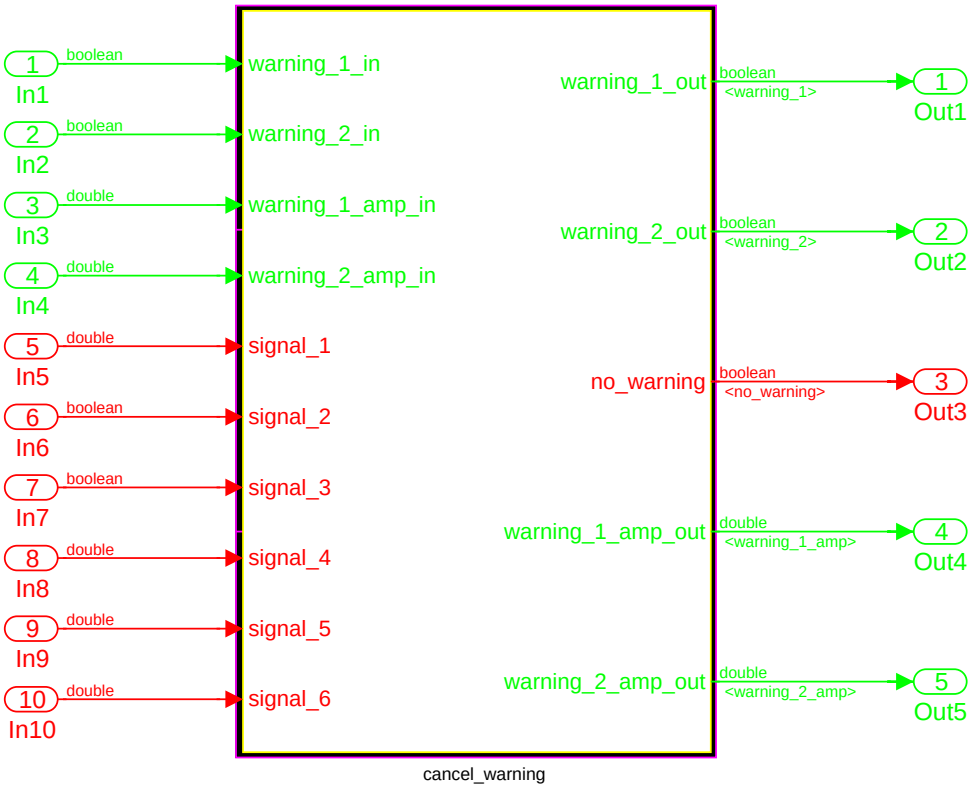



**Figure 6.3:** The `cancel_warning` subsystem.

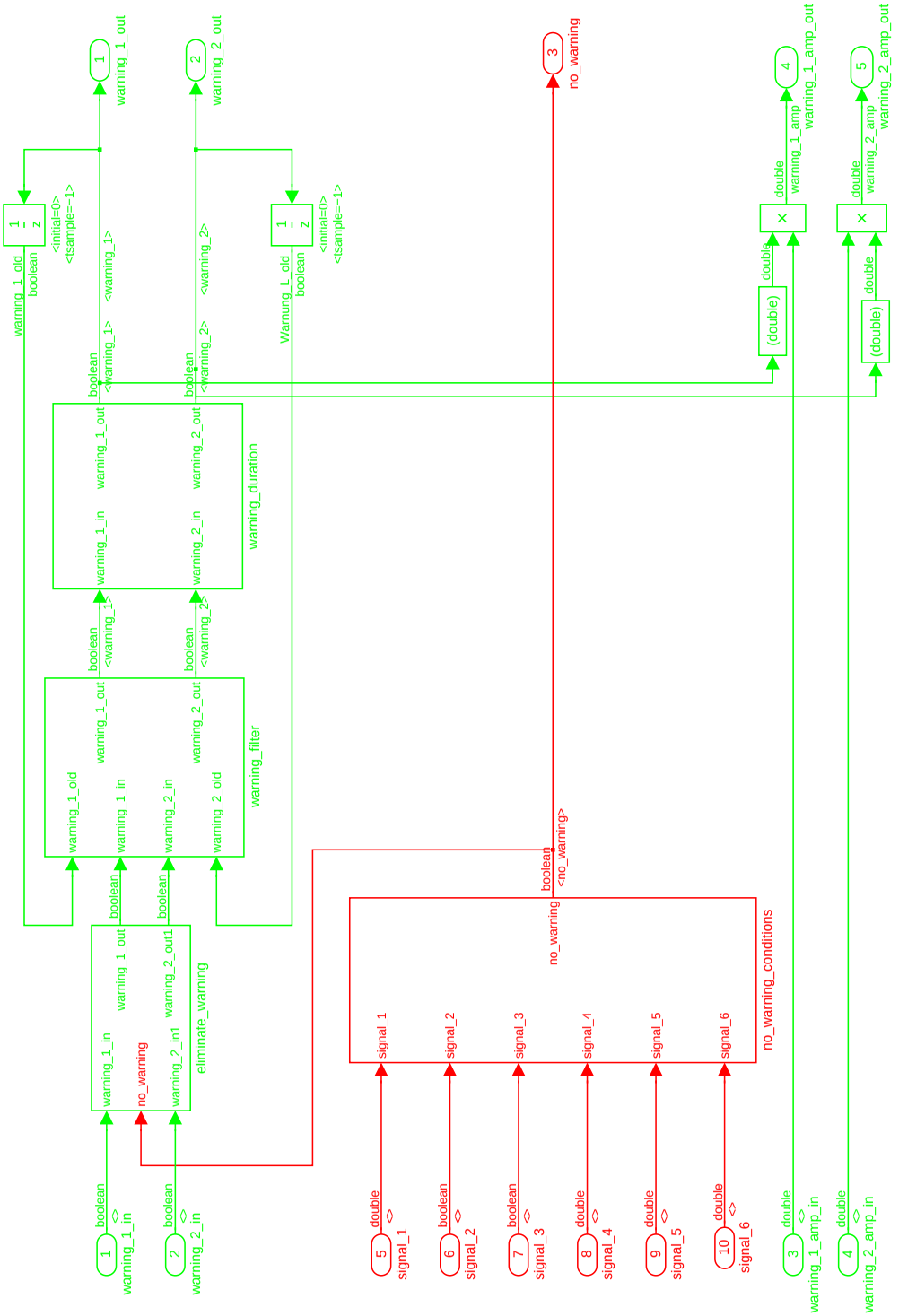



**Figure 6.4:** Contents of the `cancel_warning` subsystem.

`cancel_warning`. This node is not the main node of the LUSTRE program, but it is called from the `main` node, part of which is also shown in the figure.

Notice that the LUSTRE node `cancel_warning` contains one more input than the SIMULINK subsystem `cancel_warning`, namely, `clock1_5`. This is because of the following requirement in the LUSTRE language. When a node takes inputs not running at the basic clock of the node, the clock(s) of these inputs must be passed as argument(s) to the node. In this case, some inputs run at the basic clock of 4 ms while others run at the slower clock of 20 ms, that is, five times slower. Flow `clock1_5`, constructed in the `main` node, models this. Recall that the assumption is that the LUSTRE program will be run periodically every 4 ms. Under this assumption, flows running at the basic clock will be sampled every 4 ms (e.g., `In5`) whereas those running at the clock `clock1_5` will be sampled every 20 ms, which is what we want.

**Verification** As mentioned earlier, one of the benefits of having a formal counterpart of the SIMULINK model is the possibility for performing formal verification. We took advantage of this option and performed model-checking on the LUSTRE program generated from the warning processing system. The properties to be checked were provided by Audi, as part of our collaboration in the "NEXT TTA" project.

We used the LESAR model-checker [RHR91]. LESAR takes as input the LUSTRE program and the property to be checked, expressed in a formal way. Usually, the property is modeled in LUSTRE, using an *observer*. The observer is a LUSTRE node which outputs a boolean signal which should remain true unless the property is violated. The environment (inputs to the LUSTRE program) is also modeled, usually by *assertions*. The model checker performs exhaustive search of the state space to check if the property is violated at any reachable states. The problem is generally undecidable, since the program may contain integer and real variables. Thus, LESAR performs automatically abstraction of the infinite-domain variables (integers, reals), so that the state space is guaranteed to be finite. Abstractions are performed either using additional boolean variables or using a polyhedra library [HPR97].

As a first step, we have formalized the requirements which have been provided by Audi in English. We have then checked them using LESAR. A total of 23 properties were checked and a number of them were found to be false, given the environment constraints. It turned out that the latter were not as strict as in a realistic situation, but a more detailed model of the environment was not available. In a few cases, we had to manually modify the LUSTRE program in order for the verification to succeed in finding the error, by "hand-coding" abstractions of integer and real variables.

### 6.2.2 Steer-by-wire controller

The second model provided by Audi is part of an assistant system for a prototype car which helps the driver keep the car on the road. The system consists of a camera, a steering actuator and a networked embedded system. The latter is based on the *time triggered architecture* (TTA) and its implementation in the TTP protocol [Kop97, KG94]. In this system, four computers are used, one dedicated to image processing the camera input, one for actuating and two running the same

```
node main( In1: bool; In2: bool; ... )
returns ( Out1: bool; Out2: bool; ... );
var  In1_5: bool; In2_5: bool; ...
let
     ( Out1, Out2, ... ) = cancel_warning( clock1_5, In1_5, In2_5, ... ) ;
     In1_5 = In1 when clock1_5 ;
     ...
     clock1_5 = make_clock(5, 0) ;
tel


node cancel_warning( clock1_5:bool;
                     warning_1_in:bool when clock1_5;
                     warning_2_in:bool when clock1_5;
                     warning_1_amp_in:real when clock1_5;
                     warning_2_amp_in:real when clock1_5;
                     signal_1:real;
                     signal_2_n:bool;
                     signal_3:bool;
                     signal_4:real;
                     signal_5:real;
                     signal_6:real )

returns             ( warning_1_out:bool when clock1_5;
                      warning_2_out:bool when clock1_5;
                      no_warning:bool;
                      warning_1_amp_out:real when clock1_5;
                      warning_2_amp_out:real when clock1_5 );
let

    ...

tel
```

**Figure 6.5:** A small fragment of the LUSTRE program generated by S2L from the warning processing system model.

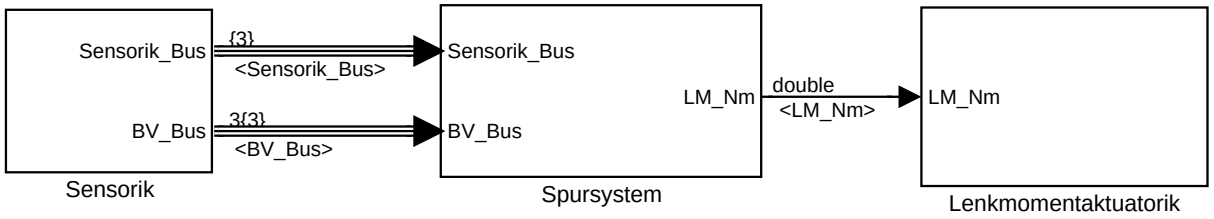



**Figure 6.6:** The steer-by-wire controller model (root system).

steering control algorithm for fault-tolerance purposes. The computers are linked with CAN bus as well as TTP network.

As part of the "NEXT TTA" project, we used S2L to automatically translate the SIMULINK model containing the control algorithm into LUSTRE. The LUSTRE program was then compiled into C using C code generators from *Esterel Technologies, Inc.*. The C code was integrated into the prototype vehicle: a minimal manual effort was required to interface the code with the TTP platform. A demonstration of the system was performed for the reviewers and project partners during the final project review on January 22, 2004 at the Audi testbed in Ingolstadt, Germany.

The root system of the SIMULINK model is shown in Figure 6.6. It consists of three subsystems. The `Sensorik` subsystem is used to receive inputs from the environment and do an initial processing. The `Spursystem` subsystem performs the main computations. The `Lenkmomentaktuatorik` subsystem is used to write the outputs. Notice that the first subsystem has no external inputs and the last one has no external outputs. This is because interfacing of the generated code is done through the use of external functions belonging in special libraries. In the SIMULINK model, inputs in the `Sensorik` subsystem are replaced by `Ground` blocks and outputs in the `Lenkmomentaktuatorik` subsystem are replaced by the `Terminator` blocks. These blocks are translated by S2L as external LUSTRE functions. The latter are then defined using external APIs and libraries.

Other parts of the model are shown in Figures 6.7 and 6.8.

The size of this model is 6 levels of hierarchy, 20 subsystems and about 150 blocks. The entire model runs at a single period. The generated LUSTRE code is 486 lines long and contains 19 nodes and 13 external functions.

### 6.2.3 A car alarm monitoring system

The next case study is the STATEFLOW model shown in Figure 6.9. This is a hypothetical alarm monitoring system for a car. This contains two parallel states, `Speedometer` which adjusts the `speed` variable according to input events and `Car` which is hierarchical, the outer layer `engine_on` monitoring the engine status and the next inner layer monitoring the car's speed. The innermost level has two parallel states, `belt` which monitors the seat belt status and generates the `belt_alarm` alarm if the seat belts are not on and the speed is greater than 10, and `locks` which monitors the door lock switch and controls the locks.

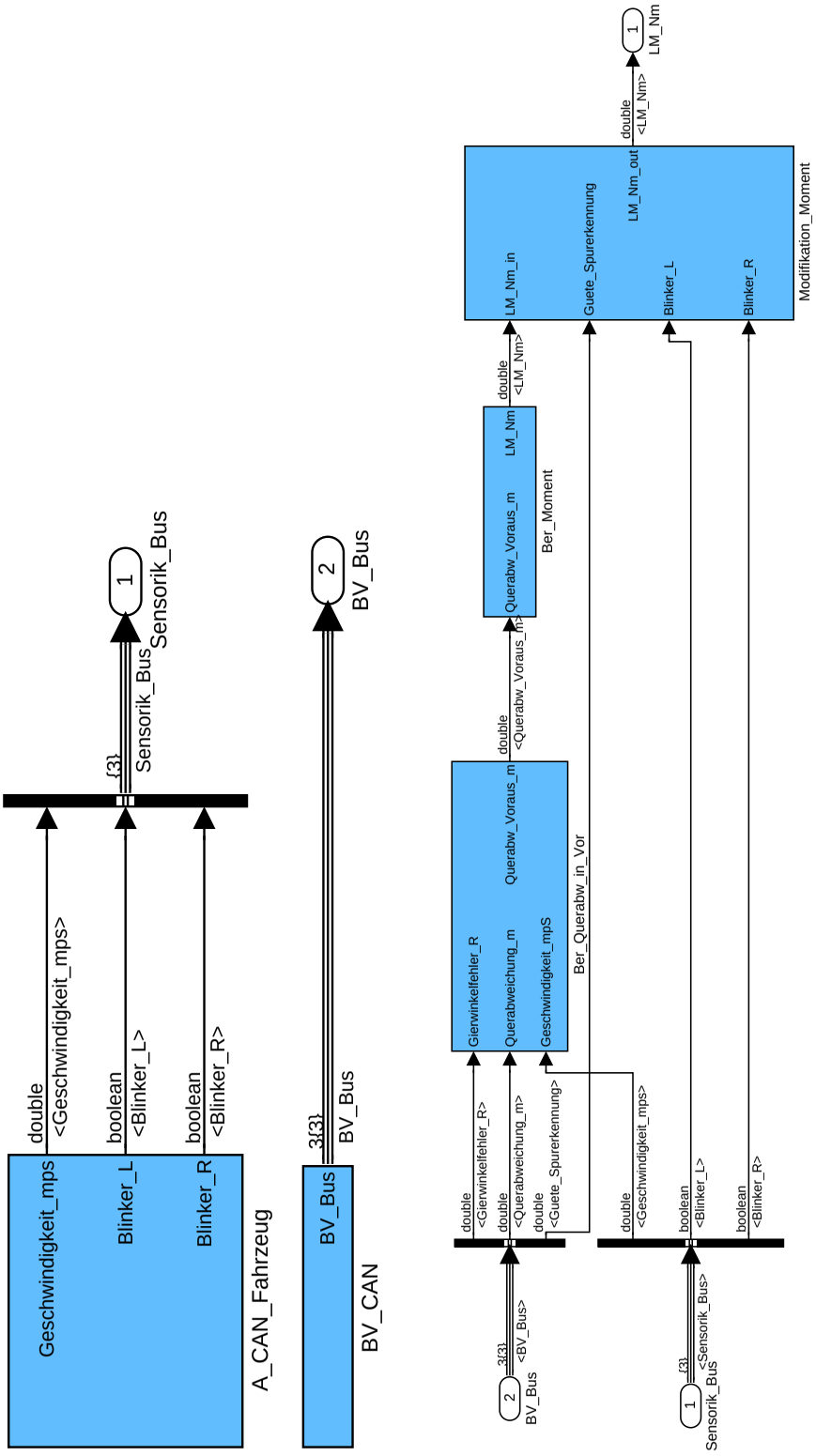



**Figure 6.7:** Parts of the steer-by-wire controller model.

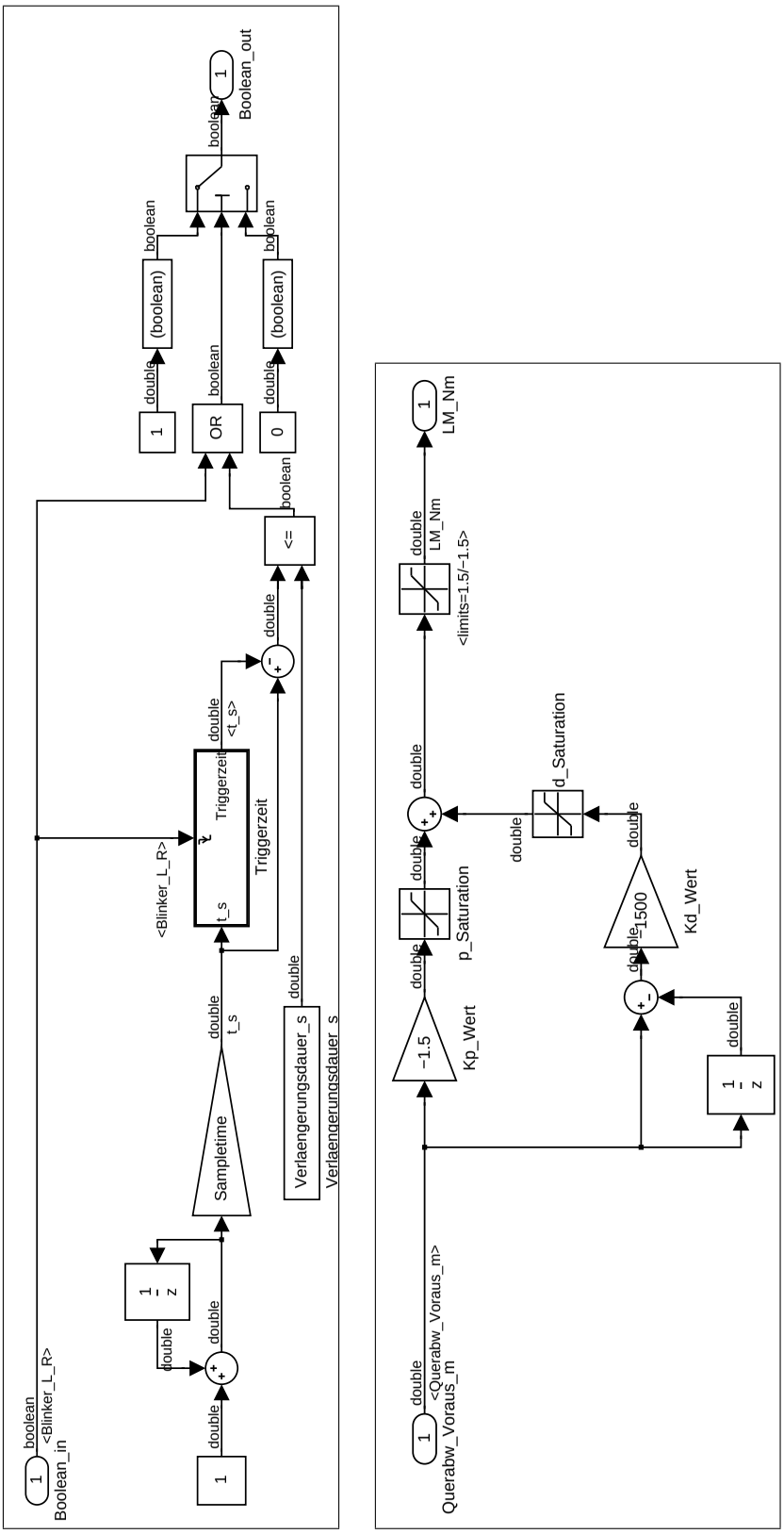



**Figure 6.8:** Parts of the steer-by-wire controller model.

LESAR only has limited support for numerical values, and does not handle the `speed` variable very well. Since we now have a LUSTRE program, we could use the tool Nbac [JHR99], which is based on abstract interpretation techniques, to handle the `speed` variable. However, the only role of this variable in the model is in boolean tests so we can abstract this variable and use an equivalent set of boolean flags. This chart is shown in Figure 6.10. Here, the `Speedometer` state outputs flags according to whether the speed is zero, non-zero or greater than 10 or 20. The rest of the model has been suitably transformed. The observer for this model states that there should be no alarms when the engine is off and that the door locks should always be on when the speed is greater than 20. Furthermore, the belt alarm should be on if the speed is greater than 10 and the `belt` status is off.

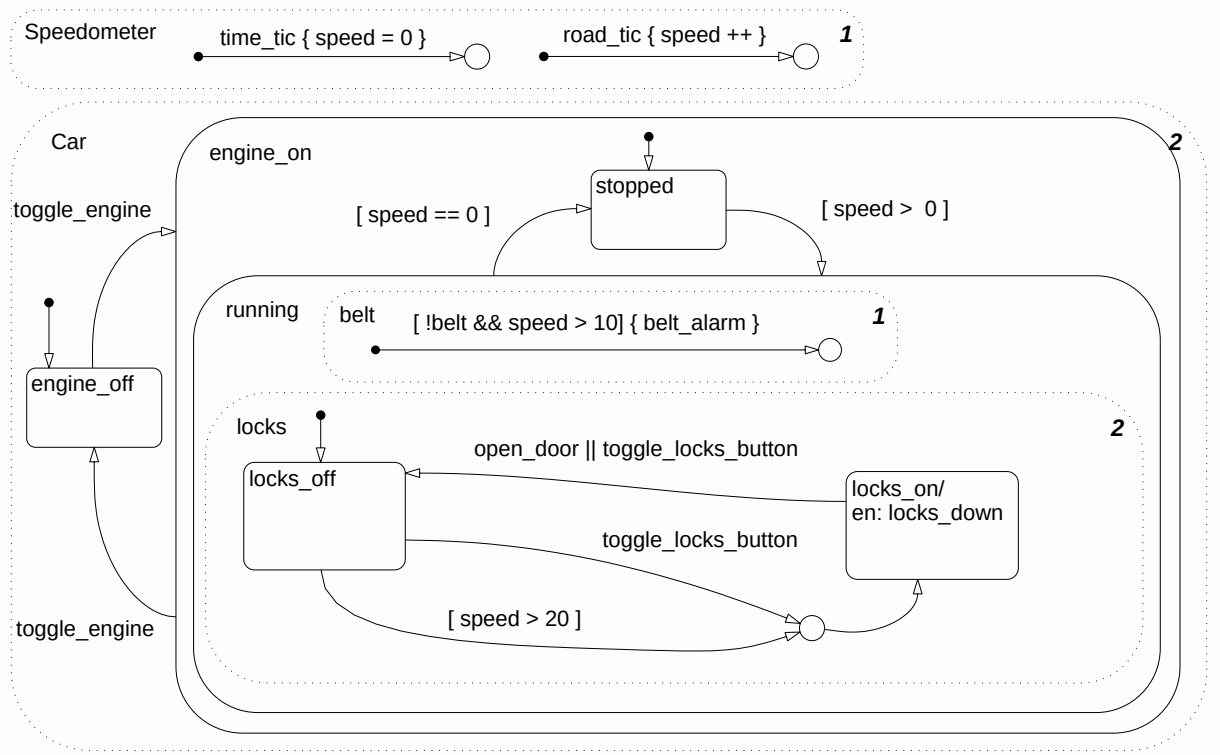



**Figure 6.9:** An alarm controller for a car

Running LESAR on the original model results in a `FALSE` property with the following counterexample:

```
--- TRANSITION 1 ---
road_tic
--- TRANSITION 2 ---
toggle_engine and not time_tic and road_tic
--- TRANSITION 3 ---
not toggle_engine and not time_tic and road_tic
```

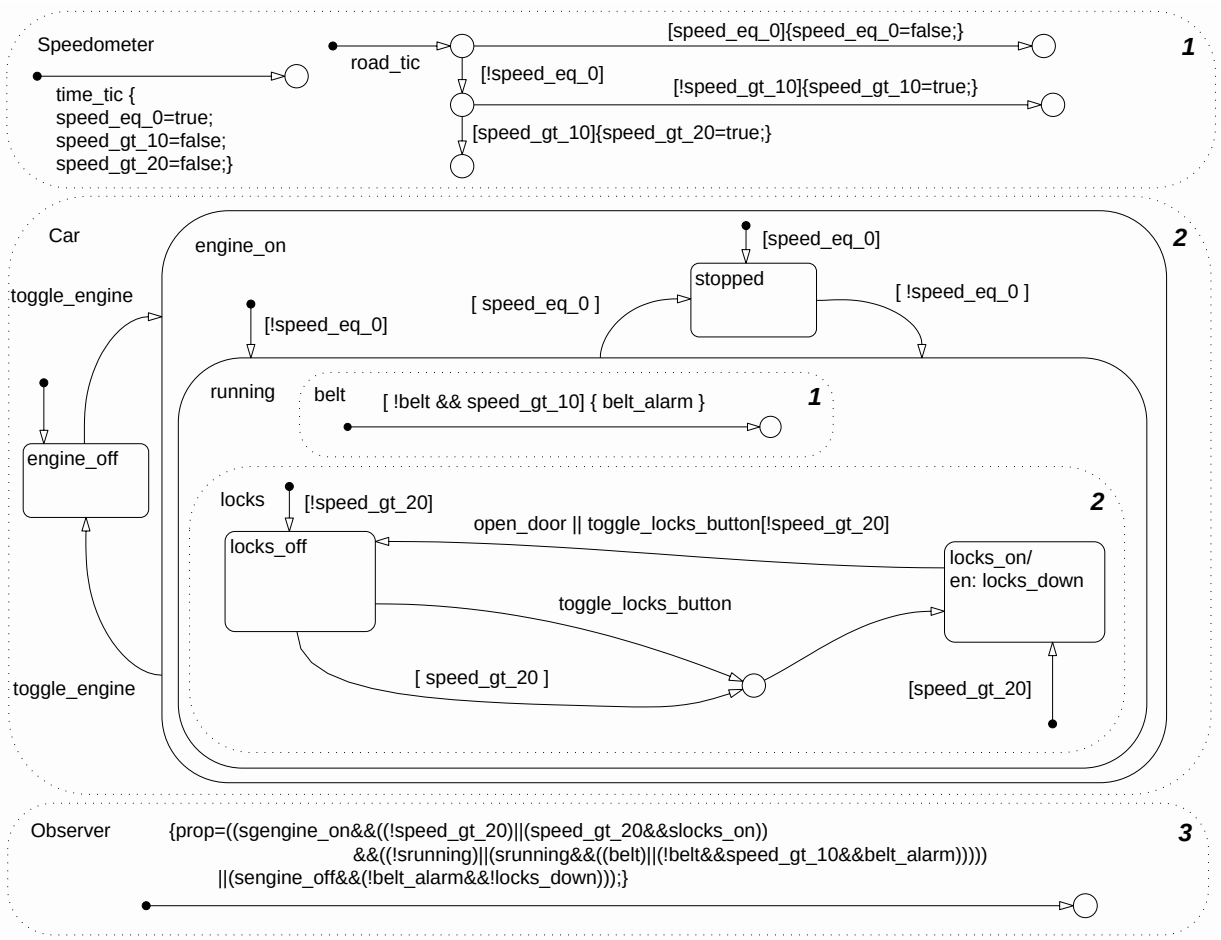



**Figure 6.10:** Abstracted and corrected version of the alarm controller

What the model-checker has spotted is that if the engine is switched on while the car is moving (not an impossibility by any means) then it is possible to reach a state where the speed is greater than 20 and not be in the `locks_on` state. The solution is simple, split up the default transitions in the `engine_on` and `locks` states (for example, `[speed_eq_0]` and `[!speed_eq_0]`) so that the correct state is reached depending upon the initial conditions when these states are entered. These additional default transitions are shown in Figure 6.10. The new model gives a `TRUE` LESAR property with the observer shown.

This model is perhaps not a realistic application but even with such a simple model the properties verified by LESAR are not intuitively obvious. It is also not very well-written STATEFLOW since the use of conditions on default transitions is warned against in the STATEFLOW documentation. The point, however, is that given suitable observers and verification by model-checking, even badly written STATEFLOW can be used with confidence.

### 6.2.4 A mixed SIMULINK/STATEFLOW case study

The last case study we present is a model that combines both a SIMULINK part and a STATEFLOW part. One of the users of our distributed platform is the group in Turku Center for Computer Science in Finland. They have used our translator, and its model checking capabilities, for their research in combining SIMULINK/STATEFLOW with *mode-automata*.

This model has been kindly provided to us by Pontus Boström of the Department of Computer Science of Abo Akademi, Finland. It is a part of the model used in [BM06] to investigate the suitability of a formalization proposed in that work.

The model represents a digital hydraulics system. The system consists of a hydraulic cylinder that moves a load mass either to a desired position or with a desired speed. The speed of the load mass is controlled by the pressure on each side of the piston in the cylinder. A digital controller controls the pressures in the cylinder using a system of on/off valves. The SIMULINK model is illustrated in Figure 6.11 and the STATEFLOW part is the `Mode switching` block in the middle of the SIMULINK diagram. The STATEFLOW controller (shown in Figure 6.12) has the following three modes *stopped*, *extending* and *retracting*.

Using our tool-chain, we managed to translate this model to LUSTRE and then to ensure that the invariant in each mode is maintained by the transitions. The initial model contains more than 50 Blocks and the hierarchy is not very deep composed of only of the two subsystems we see in Figure 6.11 as long as with one STATEFLOW Chart. The translated LUSTRE code has a length of 400 lines and contains as many as 30 LUSTRE nodes.

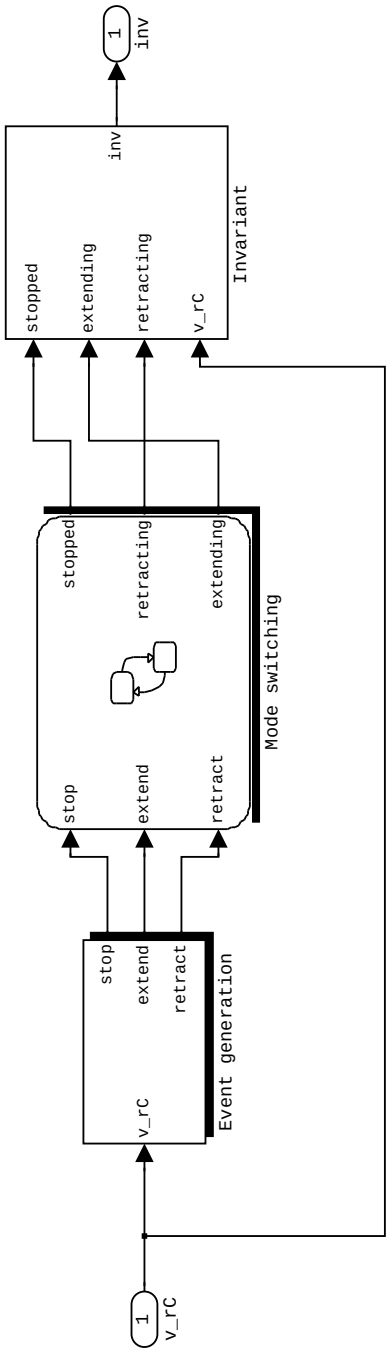



**Figure 6.11:** The SIMULINK part of the kiiku_verification example

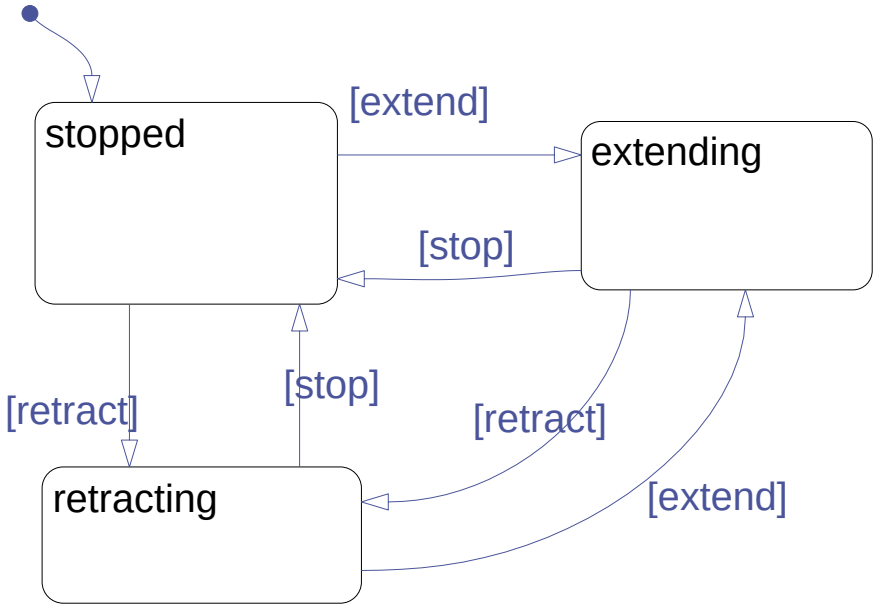



**Figure 6.12:** The STATEFLOW part of the kiiku_verification example

# Chapter 7

# Preservation of Synchronous Semantics under Preemptive Scheduling

## 7.1 Motivation

In this Chapter we study the problem of code generation from synchronous high-level languages such as SIMULINK or LUSTRE to a single-processor, multi-tasking architecture. This is in the context of the tool-chain presented in Chapter 2 (Figure 2.1).

Before we enter into the study of this problem, we should motivate the need for considering this type of architectures, and multi-tasking code generation in particular. After all, single-processor code generation methods for synchronous models exist, and in fact are the oldest: the classic single-task code generation methods (see Section 3.2 and [HCRP91]). These methods generate the simplest kind of code, that presents a number of advantages. For instance, it does not require a real-time operating system (RTOS) with multi-tasking capabilities (i.e., a scheduler to allocate the processor to the different tasks). Why then worry about multi-tasking implementations?

To understand this, let us examine more in detail the difference between the single-task and the multi-task implementation, from where we will see the motivation for the work presented in this Chapter. It is true that the generation of a single, monolithic program out of a synchronous program is simple as we saw in Section 3.2, but there is a major drawback, mostly when the synchronous program has parts running in different periods. In this case, the single-task implementation may fail the schedulability check, whereas the counterpart multi-tasking one will succeed.

We demonstrate the above using a simple example. Consider a synchronous program consisting of two parts, or tasks, $P_1$ and $P_2$, that must be executed every (, i.e., their periods are) 10 ms and every 100 ms, respectively. Suppose the worst-case execution time (WCET) of $P_1$ and $P_2$ is 2 ms and 10 ms, respectively, as shown in Figure 7.1. Then, generating a single task $P$ that includes the code of both $P_1$ and $P_2$ would be problematic. $P$ would have to be executed every 10 ms, since this is required by $P_1$. Inside $P$, $P_2$ would be executed only once every 10 times (e.g., using an internal counter modulo 10). Assuming that the WCET of $P$ is the sum of the

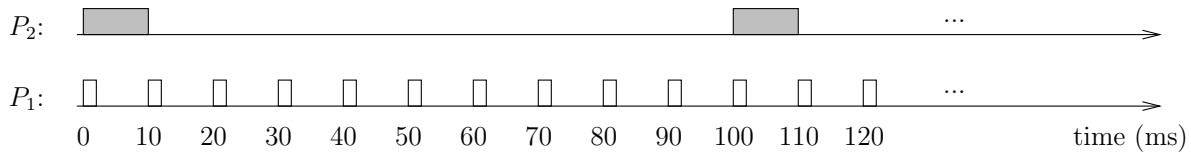



**Figure 7.1:** Two periodic tasks.

WCETs of $P_1$ and $P_2$ (note that this is not always the case), we find that the WCET of $P$ is 12 ms, that is, greater than its period. In practice, this means that every ten times, the task $P_1$ will be delayed by 2 ms. This may appear harmless in this simple case, but the delays might be larger and much less predictable in more complicated cases.

Until recently, there has been no rigorous methodology for handling this problem. In the absence of such a methodology, industrial practice consists in "manually" modifying the synchronous design, for instance, by "splitting" tasks with large execution times, like task $P_2$ above. Clearly, this is not satisfactory as it is both tedious and error-prone.

Multi-task implementations can provide a remedy to schedulability problems as the one discussed above. When an RTOS is available, and the two tasks $P_1$ and $P_2$ do not communicate with each other (i.e., do not exchange data in the original synchronous design), there is an obvious solution: generate code for two *separate* tasks, and let the RTOS handle the scheduling of the two tasks. Depending on the scheduling policy used, some parameters need to be defined. For instance, if the RTOS uses a static-scheduling policy, as this is the case most of the times, then a priority must be assigned to each task prior to execution. During execution, the highest-priority task among the tasks that are ready to execute is chosen. In the case of *multi-periodic* tasks, as in the example above, the *rate-monotonic* assignment policy is known to be optimal in the sense of *schedulability* [LL73]. This policy consists in assigning the highest priority to the task with the highest rate (i.e., smallest period), the second highest priority to the task with the second highest rate, and so on.

This solution is simple and works correctly as long as the tasks do not communicate with each other. However, this is not a common case in practice. Typically, there will be data exchange between tasks of different periods. In such a case, some inter-task communication mechanism must be used. On a single processor, this mechanism usually involves some form of *buffering*, that is, shared memory accessed by one or more tasks. Different buffering mechanisms exist:

- simple ones, such as a buffer for each pair of writer/reader tasks, equipped with a *locking* mechanism to avoid corruption of data because of simultaneous reads and writes.

- same as above but also equipped with a more sophisticated protocol, such as a *priority inheritance* protocol to avoid the phenomenon of *priority inversion* [SRL90], or a *lock-free* protocol to avoid blocking upon reads or writes [CB97, HPS02].

- other shared-memory schemes, like the *publish-subscribe* scheme used in the PATH project [PV95, Tri02], which allows *decoupling* of writers and readers.

None of these buffering schemes, however, guarantees preservation of the original synchronous semantics[1]. This means that the sequence of outputs produced by some task at the implementation may not be the same as the sequence of outputs produced by the same task at the original synchronous program. Small discrepancies between semantics and implementation can sometimes be tolerated, for instance, when the task implements a *robust* controller which has built-in mechanisms to compensate for errors. In other applications, however, such discrepancies may result in totally wrong results, with catastrophic consequences. Having a method that guarantees equivalence of semantics and implementation is then crucial. It also implies that the effort spent in simulation or verification of the synchronous program need not be duplicated for the implementation. This is an extremely important cost factor and hereafter we provide standard methods and algorithms to overcome this problem and preserve the semantics in a communication between tasks in a preemptive scheduling execution platform.

As we show later in Section 7.3.3, ”naive” inter-process communication schemes do not preserve the synchronous semantics. In particular, such schemes are not deterministic: depending on the execution time of the tasks, the data sent from a task to another might be different.

One might say that strict preservation of the synchronous semantics is not really necessary. After all, in control applications controllers are usually designed to be robust to various types of data variability, including data loss, jitter, sensor inaccuracies, etc. Two answers can be given to this claim. First, controllers contain more and more ”discrete logic”, which is not robust (a single bit-flip may change the course of an if-then-else statement). Second, echos from the industry indicate that determinism is an important requirement. For instance, recent versions of the SIMULINK code-generator REAL-TIME WORKSHOP provide options to ”ensure deterministic data transfer” (see the ”Related work” section for references). Our contacts with Esterel Technologies reveal similar concerns.

### Decomposition of a synchronous program into tasks

In the rest of this chapter we will be assuming an abstract model of a synchronous program based on a set of communicating *tasks* where each task has its own *triggering event* (see Section 7.2). The question arises, then, how to go from a synchronous model such as SIMULINK or LUSTRE to this abstract model of tasks. In other words, how to decompose a synchronous program into a set of tasks.

The decomposition procedure can be quite complicated involving WCET (worst-case execution time) estimation and schedulability analysis (i.e., checking that the tasks meet their deadlines, given a specified scheduling policy). Decomposition is indeed a topic on its own, and we do not attempt to cover it in this work. Instead, we discuss a simple decomposition method, often encountered in practice. Clearly this is a topic for future research.

The simple method boils down to grouping together all the different parts of the code that are triggered by the same event, into separate tasks, one task for each such event. This event

[1] Many of these schemes guarantee a *freshest-value* semantics, where the reader always gets the latest value produced by the writer. Freshness is desirable in some cases, in particular in control applications, where the more recent the data, the more accurate they are.

may be a periodic one (i.e., the "tick" of a periodic clock) or an aperiodic one (e.g., linked to the crank-shaft angle in an engine-control application).

We illustrate the above decomposition method on an example. Consider the SIMULINK model in Figure 6.4, Section 6.2.1. This system has two different periods: red blocks and signals have a period of 4 milliseconds (ms) and green ones have a period of 20 ms.

In this case, we can simply decompose this system in two tasks: the "red" task with period 4 ms will contain all red blocks, and the "green" task with period 20 ms will contain all the green blocks. In this example the red task communicates data to the green task, but not vice-versa. Notice that if the green task also needed to communicate data to the red one, there would be a problem, namely, a dependency cycle, at those instants where both tasks are active (i.e., every 20 ms). In such cases this simple decomposition method does not work, and more refined methods are necessary.

## 7.2 An inter-task communication model

We consider a set of *tasks*, $\mathcal{T} = \{\tau_1, \tau_2, ...\}$. The set need not be finite, which allows the modeling of, for example, dynamic creation of tasks.

To model inter-task communication, we consider a set of *data-flow links* of the form $(i, j, p)$, with $i, j \in \{1, 2, ...\}$ and $p \in \{-1, 0\}$. If $p = 0$ then we write $\tau_i \rightarrow \tau_j$, otherwise, we write $\tau_i \stackrel{-1}{\rightarrow} \tau_j$. The tasks and links result in what we shall call a *task graph*. For each $i, j$ pair, there can only be one link, so we cannot have both $\tau_i \rightarrow \tau_j$ and $\tau_i \stackrel{-1}{\rightarrow} \tau_j$.

Intuitively, a link $(i, j, p)$ means that task $\tau_j$ receives data from task $\tau_i$. If $p = 0$ then $\tau_j$ receives the last value produced by $\tau_i$, otherwise, it receives the one-before-last value (i.e., there is a "unit delay" in the link from $\tau_i$ to $\tau_j$). In both cases, it is possible that the first time that $\tau_j$ occurs[2] there is no value available from $\tau_i$ (either because $\tau_i$ has not occurred yet, or because it has occurred only once and $p = -1$). To cover such cases, we will assume that for each task $\tau_i$ there is a *default output* value $y_0^i$. Then, in cases such as the above, $\tau_j$ uses this default value.

Notice that links model data-flow, and not *precedences* between tasks.

We allow for cycles in the graph of links, provided these cycles are not *zero-delay*, that is, provided there is at least one link $(i, j, -1)$ in every cycle. Notice that we could allow zero-delay cycles if we made an assumption on the arrival patterns of tasks, namely, that all tasks in a zero-delay cycle cannot occur at the same time. However, it is often the case that tasks occur at the same time. For instance, two periodic tasks with the same initial phase, will "meet" at multiples of the least common multiplier of their periods.

### Synchronous, zero-time semantics

We associate with this model an "ideal", *zero-time* semantics. For each task $\tau_i$ we associate a set of *occurrence times* $T_i = \{t_1^i, t_2^i, ...\}$, where $t_k^i \in R_{\geq 0}$ and $t_k^i < t_{k+1}^i$ for all $k$. Because of the

[2] As we shall see shortly, we define an "ideal" zero-time semantics where a task executes and produces its result as the same time it is released. We can thus say "task $\tau_i$ occurs".

zero-time assumption, the occurrence time captures the release, start and finish times of a task. The *release* time refers to the time the task becomes ready for execution. The *start* time refers to the time the task starts execution. The *end* time refers to the time the task finishes execution. In the next section, we will distinguish these three times.

We make no assumption on the occurrence times of a task. This allows us to capture all possible situations, namely, where a task is *periodic* (i.e., released at multiples of a given period) or where a task is *aperiodic* or *sporadic*. Also note that for two tasks $i$ and $j$, we might have $t_k^i = t_m^j$, which means that $i$ and $j$ may occur at the same time. The absence of zero-delay cycles ensures that the semantics will still be well-defined in such a case.

Given time $t \geq 0$, we define $n_i(t)$ to be the number of times that $\tau_i$ has occurred until $t$, that is:

$$n_i(t) = |\{t' \in T_i \mid t' \leq t\}|.$$

We denote inputs of tasks by $x$'s and outputs by $y$'s. Let $y_k^i$ denote the output of the $k$-th occurrence of $\tau_i$. Given a link $\tau_i \rightarrow \tau_j$, $x_k^{i,j}$ denotes the input that the $k$-th occurrence of $\tau_j$ receives from $\tau_i$. The ideal semantics specifies that this input is equal to the output of the last occurrence of $\tau_i$ before $\tau_j$, that is:

$$x_k^{i,j} = y_\ell^i, \text{ where } \ell = n_i(t_k^j).$$

Notice that if $\tau_i$ has not occurred yet then $\ell = 0$ and the default value $y_0^i$ is used.

If the link has a unit delay, that is, $\tau_i \xrightarrow{-1} \tau_j$, then:

$$x_k^{i,j} = y_\ell^i, \text{ where } \ell = \max\{0, n_i(t_k^j) - 1\}.$$

Some examples of the ideal semantics are provided in the next section, where we also show potential problems that arise during implementation.

## 7.3 Execution on static-priority or EDF schedulers

### 7.3.1 Execution under preemptive scheduling policies

We consider the situation where tasks are implemented as stand-alone processes executing on a single processor equipped with a real-time operating system (RTOS). The RTOS implements a scheduling policy to determine which of the ready tasks (i.e., tasks released but not yet completed) is to be executed at a given point in time. In this work, we consider two scheduling policies:

- *Static-priority*: each task $\tau_i$ is assigned a unique *priority* $p_i$. The task with the *highest* (greatest) priority among the ready tasks executes. We assume no two tasks have the same priority, that is, $i \neq j \Rightarrow p_i \neq p_j$. This implies that at any given time, there is a unique task that may be executed. In other words, the scheduling policy is *deterministic*, in the sense that for a given pattern of release and execution times, there is a unique behavior.

- *Earliest-deadline first* or EDF: each task is assigned a unique *(relative) deadline* (in short, *deadline*). We assume no two tasks have the same deadline, that is, $i \neq j \Rightarrow d_i \neq d_j$. The task with the smallest *absolute* deadline among the ready tasks executes. The absolute deadline of a task is equal to $r + d$, where $r$ is the release time of the task and $d$ is the deadline. However, there is a case, that the scheduler cannot make a deterministic choice. This is examined thoroughly in Section 7.4.4, providing a solution and explanation using an example.

In the ideal semantics, task execution takes zero time. In reality, this is not true. A task is released and becomes ready. At some later point it is chosen by the scheduler to execute. Until it completes execution, it may be preempted a number of times by other tasks. To capture this, we distinguish, as explained above, the release time of a task $\tau_i$ from the time $\tau_i$ begins execution and from the time $\tau_i$ ends execution. For the $k$-th occurrence of $\tau_i$, these three times will be denoted $r_k^i$, $b_k^i$ and $e_k^i$, respectively.

Throughout this Chapter, we make only one assumption concerning the release times of the different tasks, namely, that the tasks are *schedulable*. Schedulability means that no task ever violates its absolute deadline, that is, that *the release of a task cannot be prior to the completion of the previous instance, if there exists any, of the same task*. Therefore, in the static-priority case, we assume that the absolute deadline is the next release time of the task, that is, the absolute deadline of the $k$-th occurrence of $\tau_i$ is $r_{k+1}^i$. In the EDF case, if $d_i$ is the deadline of $\tau_i$, then the absolute deadline of the $k$-th occurrence of $\tau_i$ is $r_k^i + d_i$.

Obviously, schedulability depends on the assumptions made on the release times and execution times of tasks. Checking schedulability is beyond the scope of this work. A large amount of work exists on schedulability analysis techniques for different sets of assumptions: see, for instance, the seminal paper of Liu and Layland [LL73], the books [HKO+93, SSRB98], or more recent schedulability methods based on timed automata model-checking [FY04]. Notice, however, that our assumption of schedulability is not related to a specific schedulability analysis method: it cannot be, since we make no assumptions on release times and execution times of tasks.

### 7.3.2 A "simple" implementation

Our purpose is to implement the set of tasks so that the ideal semantics are preserved by the implementation. It is worth examining a few examples in order to see that a "simple" implementation does not preserve the ideal semantics.

What we call simple implementation is a buffering scheme where, for each link $\tau_i \rightarrow \tau_j$, there is a buffer $B_{i,j}$ used to store the data produced by $\tau_i$ and consumed by $\tau_j$. This buffer must ensure data integrity: a task writing on the buffer might be preempted before it finishes writing, leaving the buffer in an inconsistent state. To avoid this, we will assume that the simple implementation scheme uses *atomic* reads and writes, so that a task writing to or reading from a buffer cannot be preempted before finishing.

For links with unit delays, of the form $\tau_i \stackrel{-1}{\rightarrow} \tau_j$, the simple implementation scheme uses a *double* buffer $(B_{i,j}^0, B_{i,j}^1)$. $B_{i,j}^1$ is used to store the *current* value written by the producer (i.e., the

value written by the last occurrence of the producer) and $B^0_{i,j}$ is used to store the *previous* value (i.e., the value written by the one-before-last occurrence). Every time a write occurs, the data in the buffers is *shifted*, that is, $B^0_{i,j}$ is set to $B^1_{i,j}$ and $B^1_{i,j}$ is overwritten with the new value. The reader always reads from $B^0_{i,j}$. Reads and writes are again atomic.

For the purposes of this section, we assume that each task is implemented in a way such that all reads happen right after the beginning and all writes happen right before the end of the execution of the task. Also, there is only one read/write per pair of tasks, that is, if $\tau_i \rightarrow \tau_j$ then $\tau_i$ cannot write twice to $\tau_j$. These assumptions are not part of the implementation scheme. They are a "programming style". Our aim is to show that, even when this programming style is enforced, the ideal semantics are not generally preserved. Note that these assumptions are not needed in the sections that follow: the protocols we propose work even when these assumptions do not hold. However, we will assume that every writer task writes at least once at each occurrence. This is not a restrictive assumption since "skipping" a write amounts to memorizing the previously written value (or the default output) and writing this value.

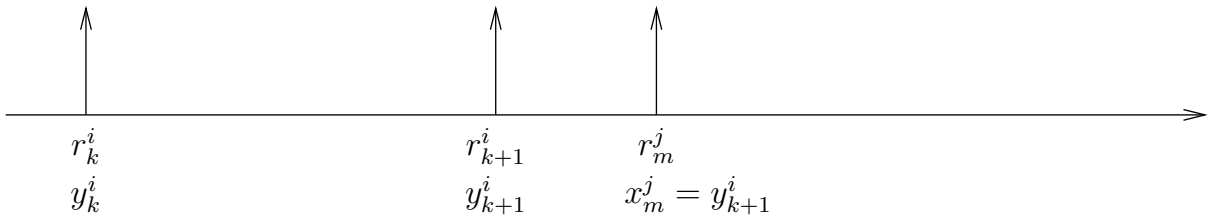
### 7.3.3 Problems with the "simple" implementation

Even with the above provisions, the ideal semantics are not always preserved. Consider, as a first example, the case $\tau_i \rightarrow \tau_j$, where static-priority scheduling is used and $\tau_i$ has lower priority than $\tau_j$, $p_i < p_j$. Consider the situation shown in Figure 7.2. We can see that, according to the semantics, the input of the $m$-th occurrence of $\tau_j$ is equal to the output of the $(k+1)$-th occurrence of $\tau_i$. However, this is not true in the implementation, because $\tau_j$ preempts $\tau_i$ before the latter has time to finish, thus, before it has time to write its result.

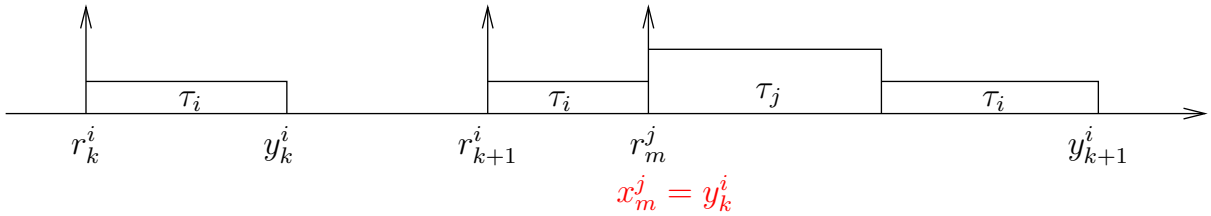
One possible solution to the above problem is to use some type of *priority-inheritance protocol*, which essentially "lifts" the priority of a task while this task is accessing a resource: see, for instance [SRL90]. In such protocols, the consumer task "blocks" and waits for the producer task to finish. In this work, we are interested in *wait-free* solutions because they are easier to implement. However, there is no wait-free solution to the above problem, unless we require that, whenever $\tau_i$ has lower priority than $\tau_j$ and $\tau_j$ receives data from $\tau_i$, a unit delay is used between the two tasks, in other words, the link must be: $\tau_i \overset{-1}{\rightarrow} \tau_j$. From now on, we will assume that this is the case.

Even when the above requirement is satisfied, the simple implementation scheme is not correct. Consider the case $\tau_i \overset{-1}{\rightarrow} \tau_j$, where $p_i < p_j$, that is, a low-to-high priority communication, with unit delay. Suppose there is another task $\tau_q$ with $p_q > p_j > p_i$. Consider the situation shown in Figure 7.3, where the order of task releases is $\tau_i$, $\tau_i$, $\tau_q$, $\tau_i$ and $\tau_j$. In the ideal semantics, the reader task $\tau_j$ uses the output $y^i_{k-1}$. However, in the simple implementation, it uses the output $y^i_{k-2}$. This is because $\tau_q$ "masks" the releases of $\tau_i$ and $\tau_j$, which results in an execution order of $\tau_i$ and $\tau_j$ which is the opposite of their arrival order.

As a third example, consider the high-to-low static-priority case, where the producer has higher priority than the consumer. In particular, we have $\tau_i \rightarrow \tau_j$ and $p_i > p_j$. There is also a third task $\tau_q$ with higher priority than both $\tau_i$ and $\tau_j$, $p_q > p_i > p_j$. Consider the situation shown in Figure 7.4. We can see that, according to the semantics, the input of the $m$-th occurrence of $\tau_j$

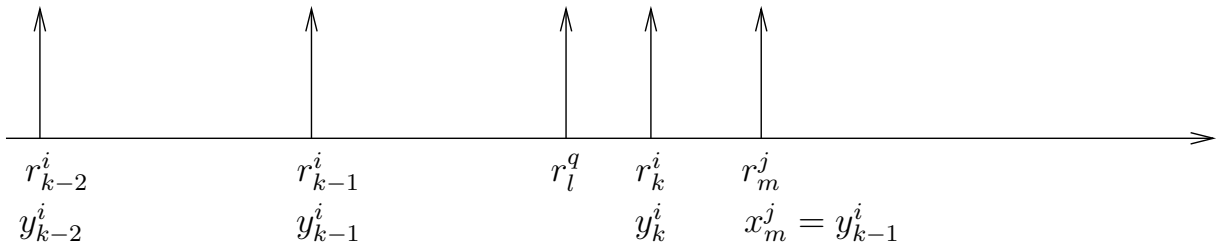



a: ideal semantics

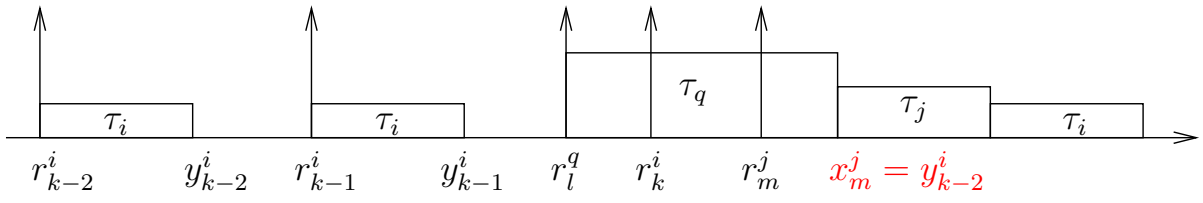



b: simple implementation

**Figure 7.2:** In the semantics, $x^j_m = y^i_{k+1}$, whereas in the implementation, $x^j_m = y^i_k$.




a: ideal semantics




b: simple implementation

**Figure 7.3:** In the semantics, $x^j_m = y^i_{k-1}$, whereas in the implementation, $x^j_m = y^i_{k-2}$.

is equal to the output of the $k$-th occurrence of $\tau_i$. However, this is not true in the implementation. This is again because $\tau_q$ "masks" the order of arrival of $\tau_j$ and $\tau_i$ ($r^j_m < r^i_{k+1}$). As a result, the order of execution of $\tau_j$ and $\tau_i$ is reversed and the reader $\tau_j$ consumes a "future" (according to the semantics) output of the writer $\tau_i$.

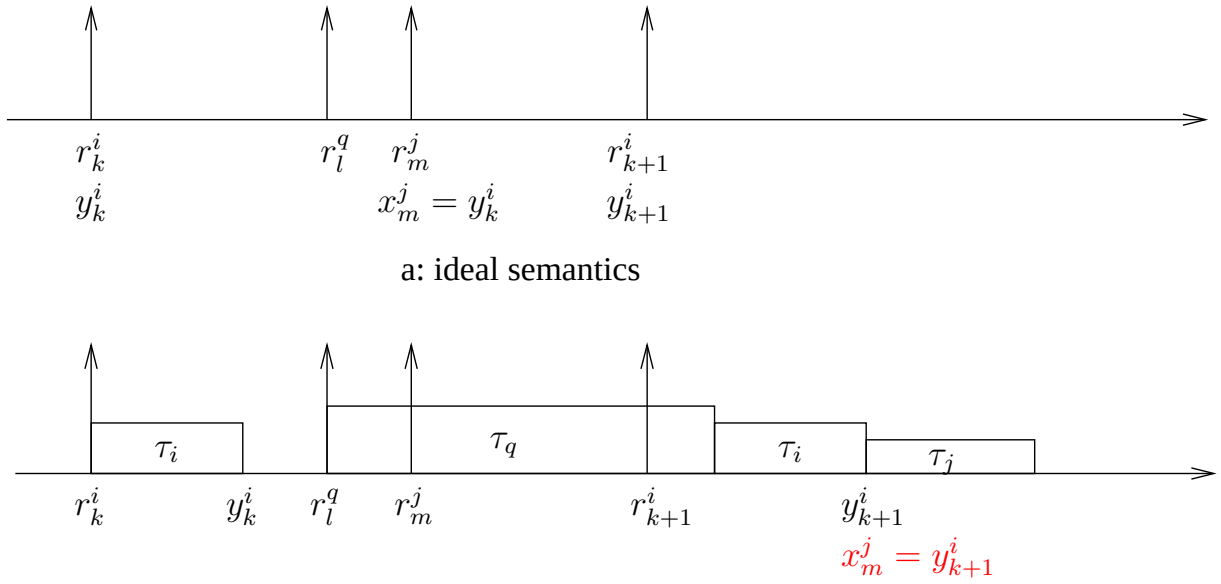



**Figure 7.4:** In the semantics, $x^j_m = y^i_k$, whereas in the implementation, $x^j_m = y^i_{k+1}$.

These examples show that a simple implementation scheme like the one above will fail to respect the ideal semantics. Note that the problems are not particular to static-priority scheduling. Similar situations can happen with EDF scheduling, depending on the deadlines of the tasks. For instance, the situation shown in Figure 7.2 can occur under EDF scheduling if $r^j_m + d_j < r^i_{k+1} + d_i$. The situation shown in Figure 7.4 can occur under EDF scheduling if $r^q_l + d_q < r^i_{k+1} + d_i < r^j_m + d_j$.

## 7.4 Semantics-preserving implementation: the one-reader case

To overcome the above problems, we propose an implementation scheme that preserves the ideal semantics. The scheme can be applied to both cases of static-priority and EDF scheduling. For simplicity and in order to facilitate understanding, we first present the scheme in the special case of a writer task communicating to a single reader task. In this case, there are three protocols depending on the relative priorities (or deadlines) of the tasks as well as on whether a unit-delay is present or not. Thus, there are three protocols: the *low-to-high protocol*, the *high-to-low*

*protocol* and the *high-to-low protocol with unit-delay*. These protocols are special cases of the general protocol presented in Section 7.5.

The first is used in the static-priority case when the writer task has lower priority than the reader task, or in the EDF case when the writer has smaller deadline than the reader. The second is used in the static-priority case when the writer has higher priority than the reader, or in the EDF case when the writer has greater deadline than the reader. The third is used in the same cases as the second, but where there is a unit-delay between the writer and the reader.

The essential idea of all protocols is that, contrary to the simple implementation scheme of the previous section, *actions must be taken not only while tasks execute but also when they are released*. These actions are simple (and inexpensive) pointer manipulations. They can therefore be provided as operating system support.

The protocols specify such release actions for both writer and reader tasks. It is essential for the correctness of the protocol that *when both writer and reader tasks are released simultaneously, writer release actions are performed before reader release actions*.

### 7.4.1 The low-to-high buffering protocol

The low-to-high buffering protocol is described in Figure 7.5. Notice that, as mentioned in the previous section, we assume a unit-delay between writer and reader. In this protocol, the writer $\tau_i$ maintains a double buffer and a one-bit variable `current`. The reader $\tau_j$ maintains a one-bit variable `previous`. `current` points to the buffer currently written by $\tau_i$ and `previous` points to the buffer written by the previous occurrence of $\tau_i$. The buffers are initialized to the default value $y_0^i$ and `current` is initialized to 0. `previous` does not need to be initialized, since it is set by the reader task upon its release.

When the writer task is released, it toggles the `current` bit. When the reader task is released, it copies the negation of the `current` bit and stores it in its local variable `previous`. Notice that these two operations happen when the tasks are *released*, and not when the tasks start executing. During execution, the writer writes to `B[current]` and the reader reads from `B[previous]`.

A typical execution scenario is illustrated in Figure 7.6. We assume static-priority scheduling in this example. One time axis is shown for each task. The double buffer is shown in the middle. The arrows indicate where each task writes to or reads from. In this example, the low-priority writer is preempted by the high-priority reader. It is worth noting that the beginning of execution of the high-priority task does not always coincide with its release. This is because, in general, there may be other tasks with even higher priority and they may delay the beginning of the task in question (in fact, they may also preempt it, but this is not shown in the figure). It can be checked that the semantics are preserved. A proof of preservation is provided in Section 7.7.

### 7.4.2 The high-to-low buffering protocol

The high-to-low buffering protocol is described in Figure 7.7. In this protocol, it is the reader $\tau_j$ that maintains a double buffer. The reader also maintains two one-bit variables `current`, `next`. `current` points to the buffer currently being read by $\tau_j$ and `next` points to the buffer

Communication: $\tau_i \overset{-1}{\rightarrow} \tau_j$.
Task $\tau_i$ maintains a double buffer `B[0,1]` and a one-bit variable `current`.
Task $\tau_j$ maintains a bit `previous`.
Initially, `current = 0` and `B[0] = B[1] =` $y_0^i$.
During execution:

- When $\tau_i$ is released: `current := not current`.
- While $\tau_i$ executes it writes to `B[current]`.
- When $\tau_j$ is released: `previous := not current`.
- While $\tau_j$ executes it reads from `B[previous]`.

Applicable under static-priority scheduling when $p_i < p_j$.
Applicable under EDF scheduling when $d_i > d_j$.

**Figure 7.5:** Low-to-high buffering protocol.

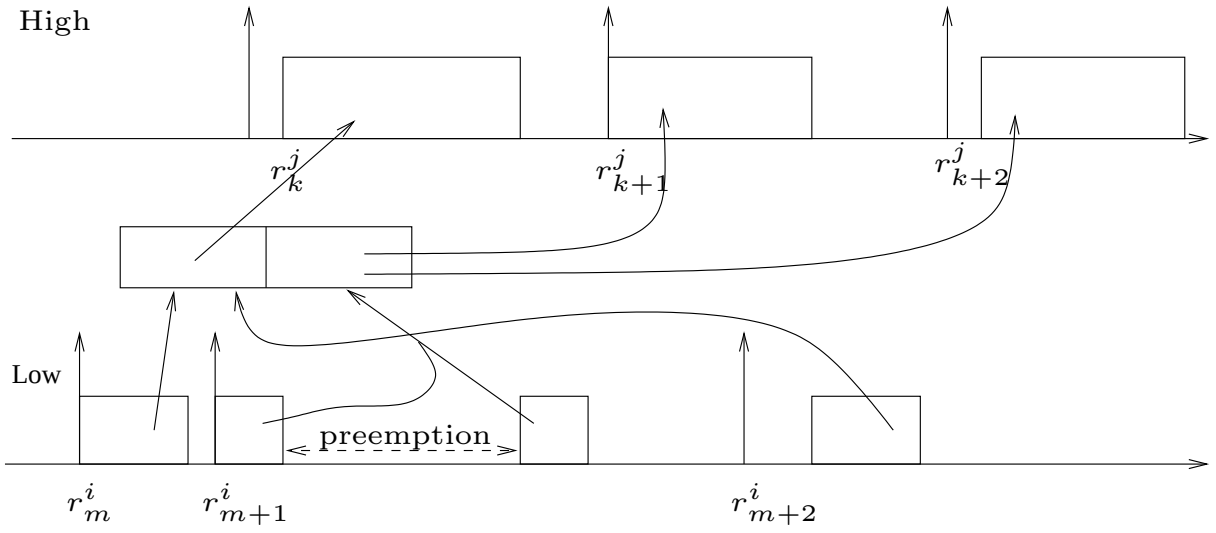



**Figure 7.6:** A typical low-to-high communication scenario.

that the writer must use when it arrives next, in case it preempts the reader. The two buffers are initialized to the default value $y_0^i$ and both bits are initialized to 0.

When the reader task is released, it copies `next` into `current`, and during its execution, it reads from `B[current]`. When the writer task is released, it checks whether `current` is equal to `next`. If they are equal, then a reader might still be reading from `B[current]`, therefore, the writer must write to the other buffer, in order not to corrupt this value. Thus, the writer toggles the `next` bit in this case. If `current` and `next` are not equal, then this means that one or more instances of the writer have been released before any reader was released, thus, the same buffer can be re-used. During execution, the writer writes to `B[next]`.

A typical execution scenario is illustrated in Figure 7.8. This example also assumes static-priority scheduling. Here, it is the reader that is preempted. It is worth noting that the beginning of execution of the high-priority writer does not coincide with its release. This is because, in general, there may be other tasks with even higher priority and they may delay the beginning of the writer. In fact, such tasks may also preempt the writer, but this is not shown in the figure.

### 7.4.3 The high-to-low buffering protocol with unit-delay

This protocol is intended for the case $\tau_i \overset{-1}{\rightarrow} \tau_j$. Before presenting this protocol, we must first point out that we can often handle this case without need for a new protocol, but using the high-to-low protocol presented above. This can be done by modifying the writer task $\tau_i$ so that it outputs not only its usual output $y^i$ but also the *previous* value of $y^i$. That is, the $k$-th occurrence of $\tau_i$ will output both $y_k^i$ and $y_{k-1}^i$, for $k = 1, 2, ....$ This can be done at the expense of adding an internal buffer to $\tau_i$, which stores the previous value of the output. Then, it suffices to "connect" the reader $\tau_j$ to $y_{k-1}^i$, which means that we have transformed the link $\tau_i \overset{-1}{\rightarrow} \tau_j$ into a link $\tau_i \rightarrow \tau_j$. Thus, we can use the high-to-low protocol of Section 7.4.2.

The modification of $\tau_i$ suggested above is not always possible, since it requires access to the task internals (e.g., source code). This is not always available, for instance, because of intellectual property restrictions. For this reason, we also provide a protocol dedicated to the high-to-low with unit-delay case. This protocol can be used by considering the writer and reader tasks as "black boxes".

The high-to-low buffering protocol with unit-delay is described in Figure 7.9. In this protocol, the reader $\tau_j$ maintains a *triple* buffer. There are also three pointers `previous`, `current` and `reading`. `previous` points to the buffer that contains the previous last value written by $\tau_i$. This is the value that was written by the execution of the one before last occurrence of the writer (the execution that correspond to the previous last release of the writer). `current` points to the buffer that the writer last wrote to or is still writing to. Finally, `reading` points to the buffer that $\tau_j$ is reading from. Buffer `B[0]` is initialized to the default value $y_0^i$. Pointers `previous` and `reading` are initialized to 0 whereas `current` is initialized to 1.

When the reader task is released, it copies `previous` into `reading`, and during its execution reads from `B[reading]`. When the writer is released, it sets `previous` to `current`, so that `previous` points to the value previously written. Then, `current` is assigned to a *free* position in the buffer, that is, a position different from both `previous` and `reading`. Note

Communication: $\tau_i \rightarrow \tau_j$.
Task $\tau_j$ maintains a double buffer `B[0,1]` and two one-bit variables `current, next`.
Initially, `current = next = 0` and `B[0] = B[1] =` $y_0^i$.
During execution:

- When $\tau_i$ is released: if `current = next`, then `next := not next`.
- While $\tau_i$ executes it writes to `B[next]`.
- When $\tau_j$ is released: `current := next`.
- While $\tau_j$ executes it reads from `B[current]`.

Applicable under static-priority scheduling when $p_i > p_j$.
Applicable under EDF scheduling when $d_i < d_j$.

**Figure 7.7:** High-to-low buffering protocol.

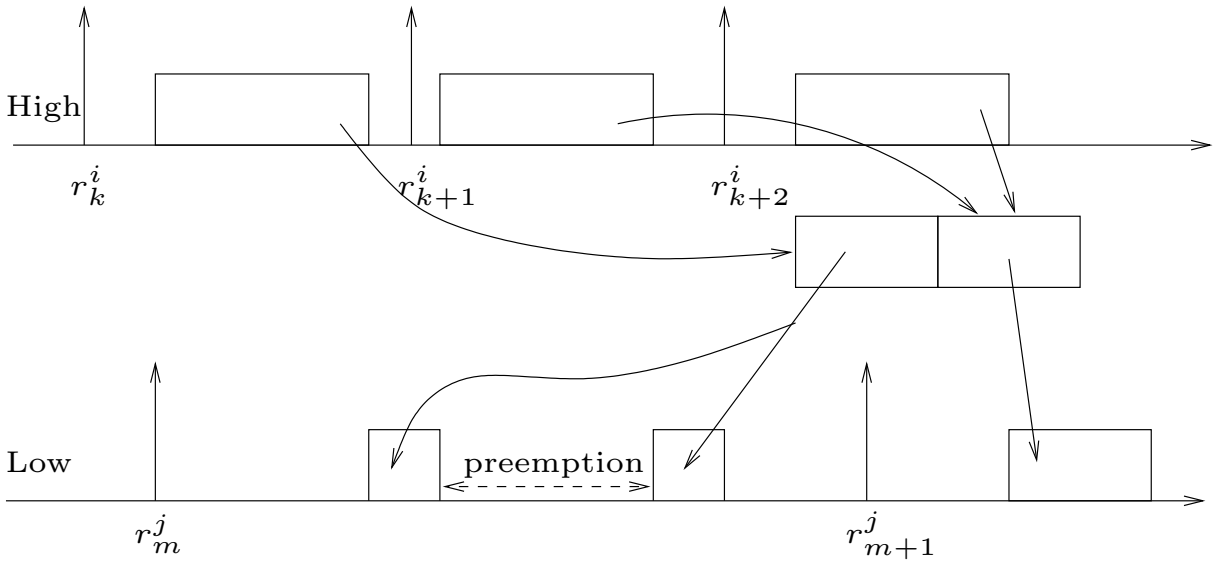



**Figure 7.8:** A typical high-to-low communication scenario.

that after the first execution the `previous` pointer will still point in the first buffer which has the default value.

### 7.4.4 Some examples under EDF scheduling

All examples we have given so far have assumed static-priority scheduling. It is probably easier for the reader to be persuaded of the correctness of the protocols in the static-priority case, before going to the other scheduling policy, the EDF.

In this section, we provide some informal arguments and examples to justify intuitively why the protocols can also be applied to EDF. Notice that, *a priori*, one might think that the protocols are not applicable to EDF, for the following reason. Under EDF, the priorities of tasks change *dynamically*. On the other hand, the above protocols have been designed assuming a *static* priority assignment to the writer and the reader. Indeed, if the two priorities are swapped, a different protocol must be used. These facts could lead one to conclude that, under EDF, the buffering scheme needs to be dynamic as well.

Fortunately, this is not the case: the buffering scheme can be defined statically, that is, before execution begins. In particular, the buffering scheme depends on the relative deadlines $d_i$ and $d_j$ of the writer $\tau_i$ and the reader $\tau_j$, respectively.

- If $d_i > d_j$ then the low-to-high buffering scheme is used. Again, we assume a unit-delay between $\tau_i$ and $\tau_j$, in order to avoid the problem of Figure 7.2.
- If $d_i < d_j$ and $\tau_i \rightarrow \tau_j$, then the high-to-low buffering scheme is used.
- If $d_i < d_j$ and $\tau_i \stackrel{-1}{\rightarrow} \tau_j$, then the high-to-low with unit-delay scheme is used.

The case $d_i > d_j$ implies that, if $\tau_j$ is released before $\tau_i$ then $\tau_i$ cannot preempt $\tau_j$, neither can it start before $\tau_i$ ends. Indeed, $r_k^j \leq r_m^i$ and $d_j < d_i$ implies $r_k^j + d_j < r_m^i + d_i$, that is, the absolute deadline of $\tau_j$ is smaller than that of $\tau_i$. Therefore, we have a situation which is "almost the same" as the low-to-high static-priority case. The difference is that in the static-priority case $\tau_j$ *always preempts* $\tau_i$, whereas in the EDF case this might not happen. Therefore, in order to guarantee the correctness of the scheme, we must examine this last possibility, to ensure that nothing goes wrong.

Figure 7.11 illustrates what might happen when $\tau_j$ does not preempt $\tau_i$ as it normally would in the low-to-high static-priority scenario. One can see that this poses no problems for the buffering scheme. In fact, the situation is as if the $k$-th instance of $\tau_j$ was released after the $(m+1)$-th instance of $\tau_i$ finished.

Let us now turn to the case $d_i < d_j$. This case implies that, if $\tau_i$ is released before $\tau_j$ then $\tau_j$ cannot preempt $\tau_i$, neither can it start before $\tau_i$ ends. Indeed, $r_k^i \leq r_m^j$ and $d_i < d_j$ implies $r_k^i + d_i < r_m^j + d_j$, that is, the absolute deadline of $\tau_i$ is smaller than that of $\tau_j$. Therefore, we have a situation which is "almost the same" as the high-to-low priority case. The difference is that in the high-to-low priority case $\tau_i$ *always preempts* $\tau_j$, whereas in the EDF case this might not happen. As before, we must examine this possibility.

Communication: $\tau_i \overset{-1}{\rightarrow} \tau_j$.
Task $\tau_j$ maintains a triple buffer `B[0..2]` and three one-bit variables `previous, current, reading`.
Initially, `previous = reading = current = 0`, `B[0]` $= y_0^i$.
During execution:

- When $\tau_i$ is released:
  `previous := current;`
  `current := x∈[0..2].(x ≠ previous ∧ x ≠ reading).`
- While $\tau_i$ executes it writes to `B[current]`.
- When $\tau_j$ is released: `reading := previous`.
- While $\tau_j$ executes it reads from `B[reading]`.

**Figure 7.9:** High-to-low buffering protocol with unit delay.

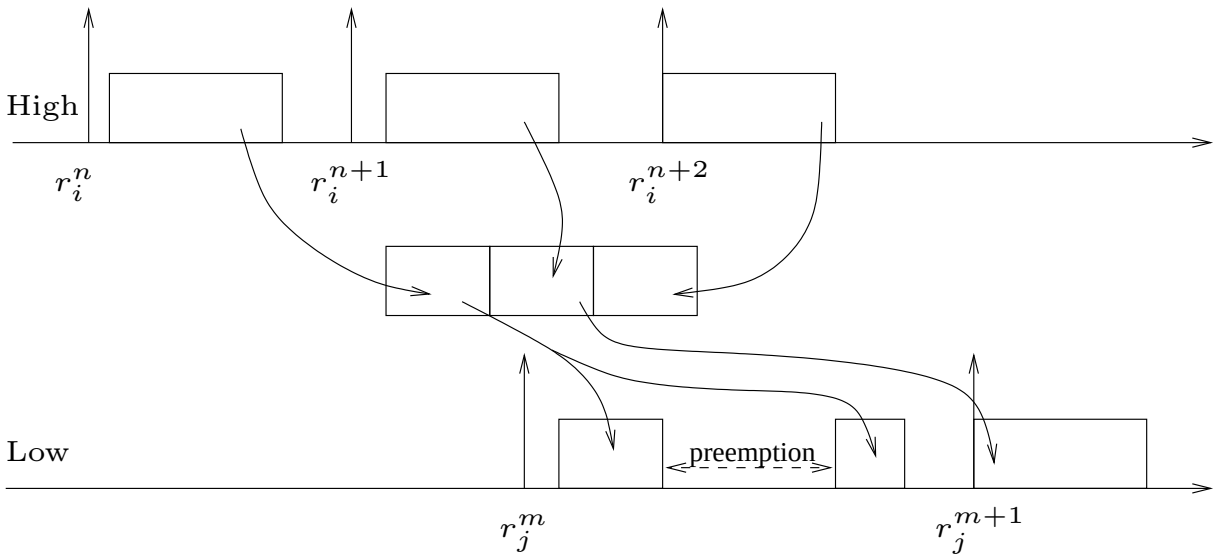



**Figure 7.10:** A typical high-to-low with unit delay communication scenario.

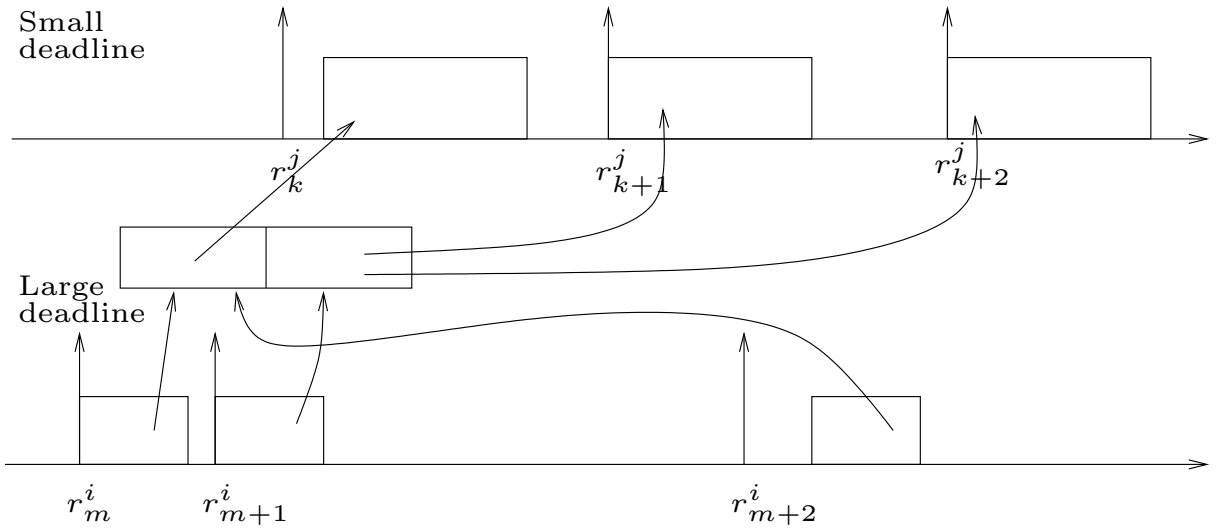



**Figure 7.11:** The scenario of Figure 7.6 possibly under EDF: $\tau_i$ is not preempted.

Figure 7.12 illustrates what might happen when $\tau_i$ does not preempt $\tau_j$ as it normally would in the high-to-low static-priority scenario. Again, this poses no problems to the buffering scheme. The situation is as if the $(k+1)$-th instance of $\tau_i$ was released after the $m$-th instance of $\tau_j$ finished.

The same situation is for the case where $\tau_i \overset{-1}{\rightarrow} \tau_j$ and $d_i < d_j$. Figure 7.13 shows a typical execution, where the writer task does not preempt the reader, despite the fact that the latter has larger deadline. This does not cause any problem, however. In fact, the situation is exactly the same as the one where $\tau_i$ arrives after $\tau_j$ finishes.

### EDF special case: equal absolute deadlines

When we were defining the task model, in Section 7.2, and the scheduling policies later, we provided means to the scheduler so that he has always a deterministic choice about the next executing task; we opposed that no two tasks can have the same priorities, for the SP, and no two tasks can have equal relative deadlines.

However, there is a case, under EDF, where the scheduler cannot distinguish between two tasks, which one is to be executed first: when the release of a task will have the same absolute deadline with the task actually running or with a task that is already preempted and waiting to continue execution.

We could bypass this problem by instructing the scheduler to continue executing the oldest task. Likewise we avoid the cost of context switching, when the new task is about to preempt an already running task. Nevertheless, we show in this section that this non-deterministic choice does not affect the preservation of semantics provided by our protocols.

We will show the above, using the example in Figure 7.14, figuring three tasks $\tau_i$, $\tau_j$ and $\tau_q$, with relative deadlines $d_q < d_j < d_i$ and task $\tau_i$ communicates its data to task $\tau_j$. Since reader

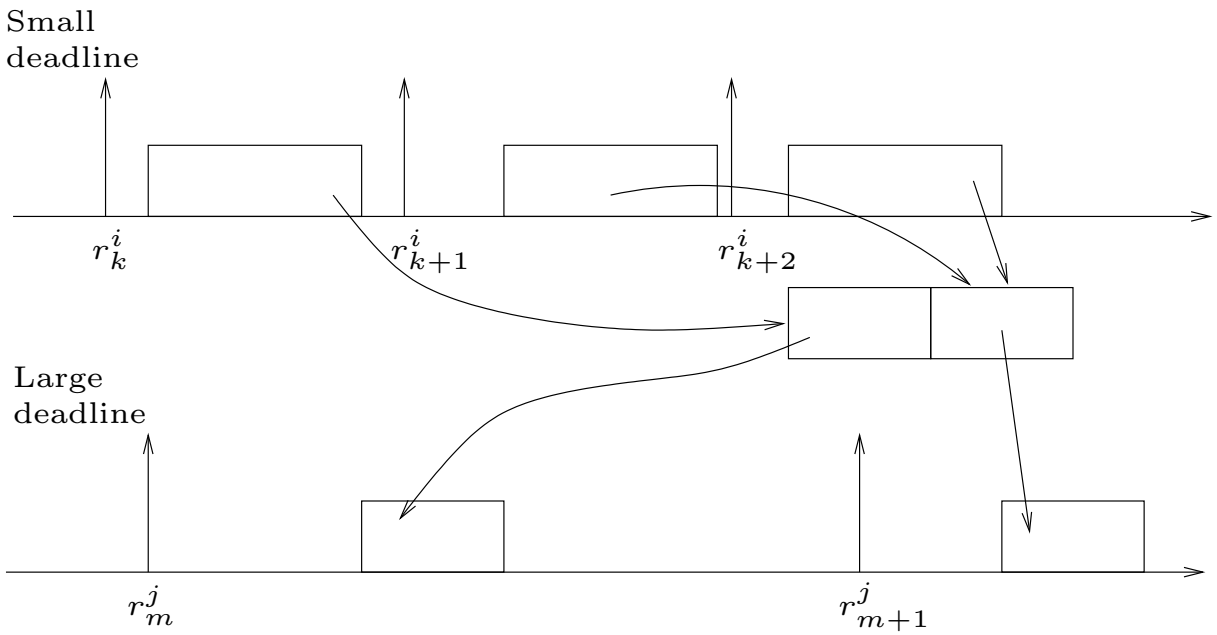



**Figure 7.12:** The scenario of Figure 7.8 possibly under EDF: $\tau_j$ is not preempted.

$\tau_j$ has shorter relative deadline than writer $\tau_i$, we will use the low-to-hi buffering protocol, as we saw in the previous section, which means that there is a *unit-delay* from the reader to the writer.

Without loss of generality, we consider that the first execution of writer task $\tau_i$ in the figure, is after a series of releases and executions of the tasks and that all of them have completed. Suppose also that `current=0` Back to the example, upon release of the writer $\tau_i$ at time $t_{r'_i}$ and according to the hi-to-low protocol, we have `current=1` and during execution $\tau_i$ writes in buffer `B[current]=B[1]`.

On the second release of the writer, at time $t_{r_i}$ the new value of the pointer is `current=0` and during its execution, $\tau_i$ writes to `B[current]=B[0]`. However, before the completion of $\tau_i$, there is the release of task $\tau_q$, at time $t_{r_q}$, whose absolute deadline $t_{r_q} + d_q$ is shorter than the one of $\tau_i$. Thus, the scheduler chooses to preempt $\tau_i$ and execute $\tau_q$. Before the end of the execution of task $\tau_q$, at time $t_{r_j}$, we have the release of the reader task $\tau_j$. Always according to the protocol, there will be new assignment for the `previous` pointer: `previous=¬current=1` and when $\tau_j$ will be executing, it will read from buffer `B[previous]=B[1]`.

The subtle point of this example, is that the arrival of the reader $\tau_j$ happens so that its absolute deadline is equal to the one of the writer $\tau_i$, already preempted. Indeed, $t_{r_i} + d_i = t_{r_j} + d_j$, as it is shown in the Figure, so when $\tau_q$ finishes execution, the scheduler will not be able to decide which task to execute next.

Our answer to this situation is that, simply, it makes no difference which will be the scheduler's choice. As a matter of fact, the two executions of Figure 7.14 demonstrate what's the series of events in any of the two choices. On top the scheduler chooses to continue the execution of the writer and after it finishes, then it executes the reader, whereas in the bottom the execution of

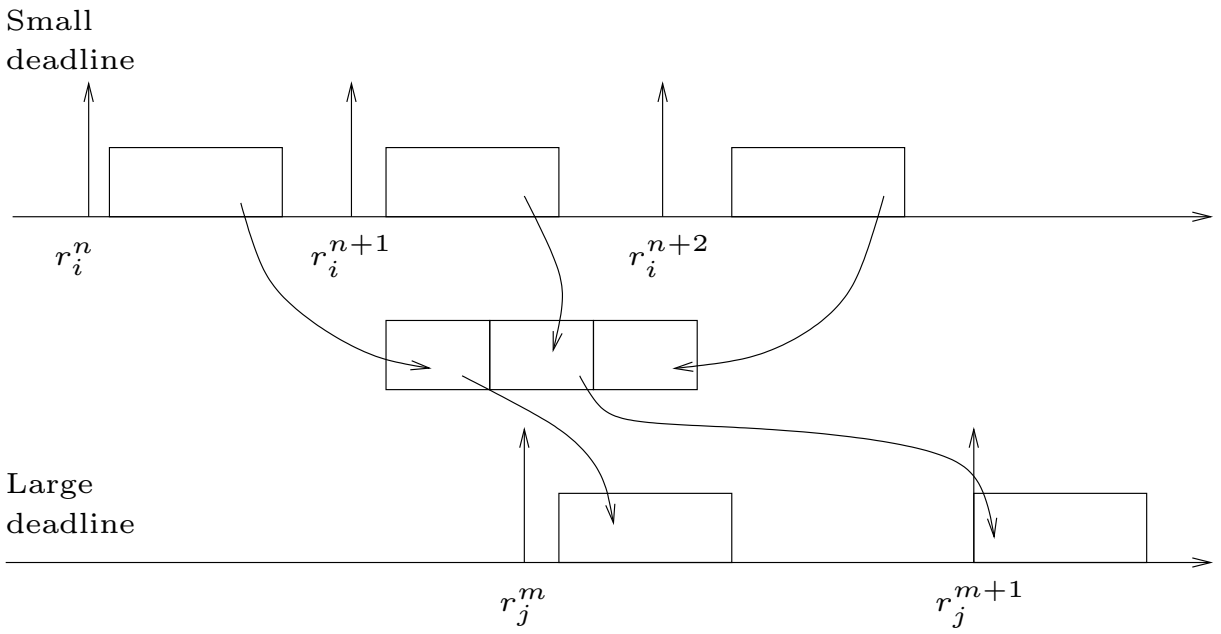



**Figure 7.13:** The scenario of Figure 7.10 possibly under EDF: $\tau_j$ is not preempted.

the reader task will preempt, for the second time, the execution of the writer.

Since there can be no more release of the reader or the writer before time $t_{r_i} + d_i$ (remember that the schedulability assumption states that there can be no new release of a task before it's absolute deadline), the pointer values `current=0` and `previous=1`, stay unchanged and thus, the writer keeps writing in `B[0]` and the reader keeps reading from `B[1]`, no matter what is their execution order, as is depicted in the both cases of the figure.

### 7.4.5 Application to general task graphs

The three protocols presented above can also be used in a general task graph as the one described in Section 7.2. Here, we show how this can be done in a simple way. Notice that the method we present here is not always optimal in terms of buffer utilization. Section 7.5 presents a generalized protocol which is also optimal.

We can assume that tasks are ordered with respect to their priorities or deadlines, depending on whether we are in a static-priority or EDF setting, respectively. For instance, in a static-priority setting, we assume that tasks are ordered as $\tau_1, \tau_2, ...$, meaning that $\tau_1$ has the highest priority, $\tau_2$ has the second highest, and so on. In an EDF setting, $\tau_1$ has the smallest deadline, and so on.

We will also assume, as we already said above, that there is no link $\tau_i \rightarrow \tau_j$ such that $i > j$. This would correspond to the low-to-high priority case without unit-delay, where semantics cannot be generally preserved. Thus, with $i > j$, we have only three types of links, namely, $\tau_i \stackrel{-1}{\rightarrow} \tau_j$, $\tau_j \rightarrow \tau_i$ and $\tau_j \stackrel{-1}{\rightarrow} \tau_i$.

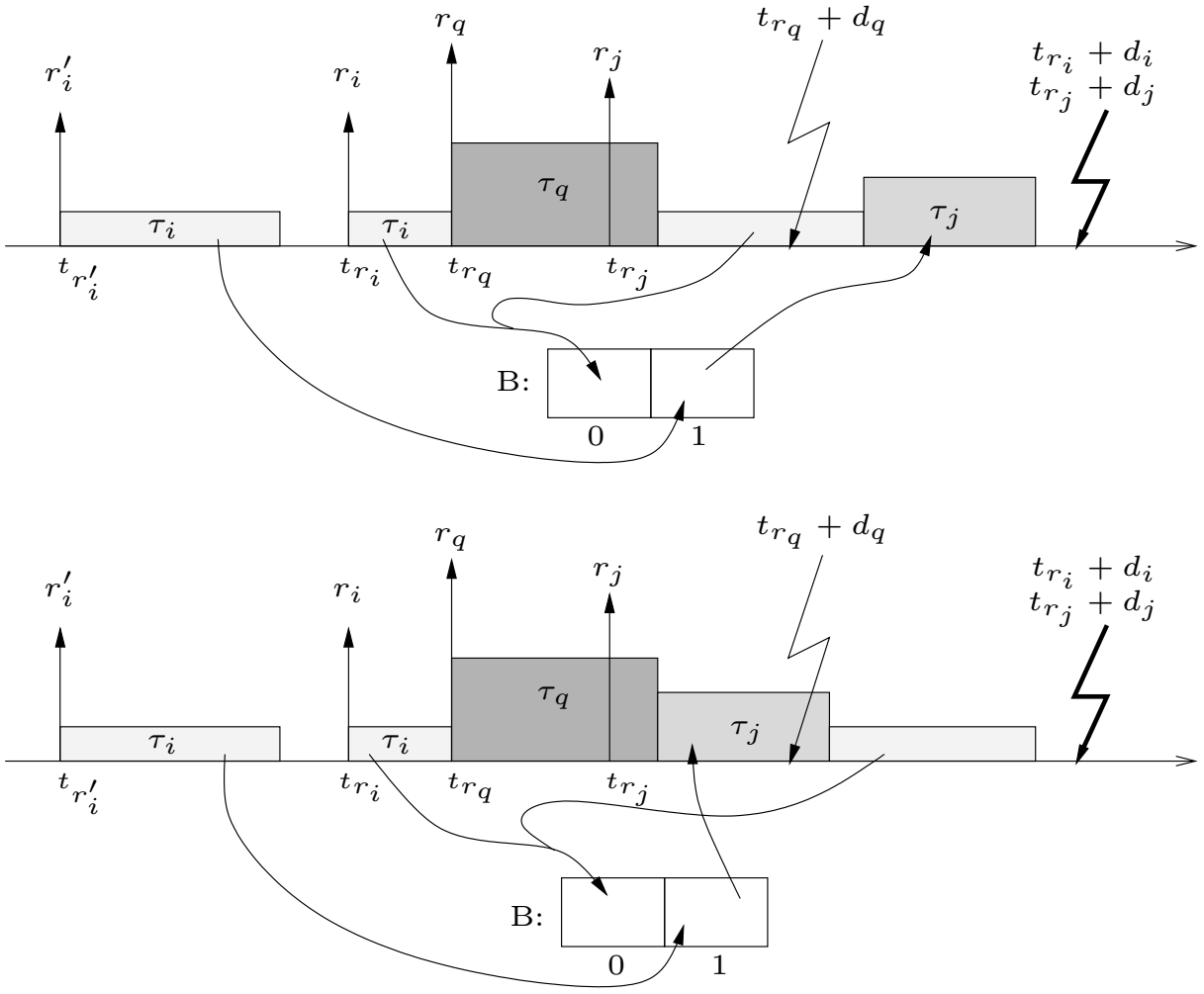



**Figure 7.14:** Equal absolute deadlines example.

One simple way of using the protocols is to consider each data-flow link separately. In other words, assuming $i > j$, for each link $\tau_i \stackrel{-1}{\rightarrow} \tau_j$ we apply the low-to-high protocol, for each link $\tau_j \rightarrow \tau_i$ we apply the high-to-low protocol, and for each link $\tau_j \stackrel{-1}{\rightarrow} \tau_i$ we apply the high-to-low protocol with unit-delay. This method results in a memory requirement of $2M + 2N_1 + 3N_2$ (single) buffers, where $M$ is the number of $\tau_i \stackrel{-1}{\rightarrow} \tau_j$ links in the task graph, $N_!$ is the number of $\tau_j \rightarrow \tau_i$ links and $N_2$ is the number of $\tau_j \stackrel{-1}{\rightarrow} \tau_i$ links. We also have memory requirements for the one-bit variables, but these are negligible. Note that we use the notation of $M$,$N_1$ and $N_2$ because it will be similar (corresponding to the case we use) when we explain the generalized protocol in Section 7.5.1.

We can immediately improve the above memory requirement by observing that, in the case of the low-to-high protocol, it is the writer that maintains the double buffer and not the reader. Therefore, if we have a set of links $\tau_i \stackrel{-1}{\rightarrow} \tau_{j_k}$ with $i > j_k$ for $k = 1, ..., M$, and if $\tau_i$ communicates the *same* data to all tasks $\tau_{j_k}$, then we do not need $2m$ buffers, but only 2.

Let us give an example. Consider the task graph shown in Figure 7.15. Unit-delays are depicted as $-1$ on the links. Notice that all unit-delays are mandatory, except the one on the link $\tau_3 \stackrel{-1}{\rightarrow} \tau_4$. We assume that every writer communicates the same data to all readers.

Using the simple method, we have buffer requirements equal to $2M + 2N_1 + 3N_2 = 2 * 4 + 2 * 2 + 3 * 1 = 15$. Using the improved method, the buffers maintained by each task are as follows:

- $\tau_1$ is the highest-priority (or lowest-deadline) task. It maintains one double buffer as the writer of the link $\tau_1 \rightarrow \tau_3$. It maintains no buffer as a reader, since in this case the buffers are maintained by the lower-priority writers.
- $\tau_2$ maintains no buffer.
- $\tau_3$ maintains one double buffer as the writer of the links $\tau_3 \stackrel{-1}{\rightarrow} \tau_1$ and $\tau_3 \stackrel{-1}{\rightarrow} \tau_2$.
- $\tau_4$ maintains one double buffer as the writer of the links $\tau_4 \stackrel{-1}{\rightarrow} \tau_1$ and $\tau_4 \stackrel{-1}{\rightarrow} \tau_2$. $\tau_4$ also maintains a triple buffer as the reader of the link $\tau_3 \stackrel{-1}{\rightarrow} \tau_4$.
- $\tau_5$ maintains one double buffer as the reader of the link $\tau_3 \rightarrow \tau_5$.

Thus, in total, the improved method uses $2 + 2 + 2 + 3 + 2 = 11$ buffers, or 4 buffers less than the simple method. There is still room for improvement, however. In particular, we show in the next section how the buffer requirements can be optimized.

## 7.5 Semantics-preserving implementation: the general case

As already mentioned, the protocols presented in Section 7.4 are specializations of a generalized protocol, called DBP, that we present in this section. DBP is used for one writer communicating (the same) data to $N$ lower-priority (or larger-deadline) readers and $M$ higher-priority (or smaller-deadline) readers, as shown in Figure 7.16. In $N_1$ among the $N$ lower-priority readers

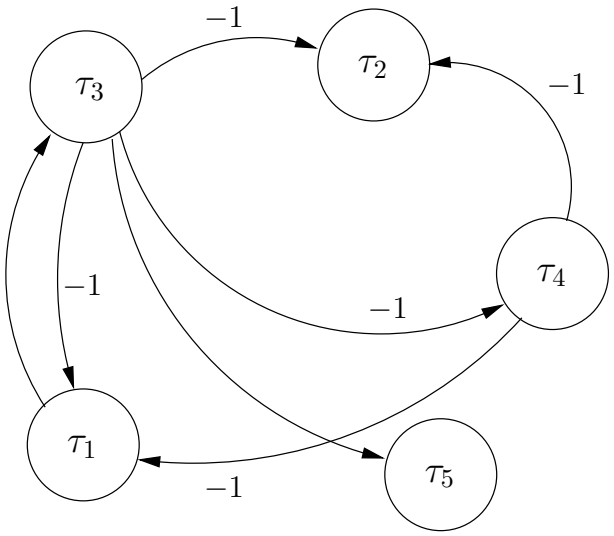



**Figure 7.15:** A task graph.

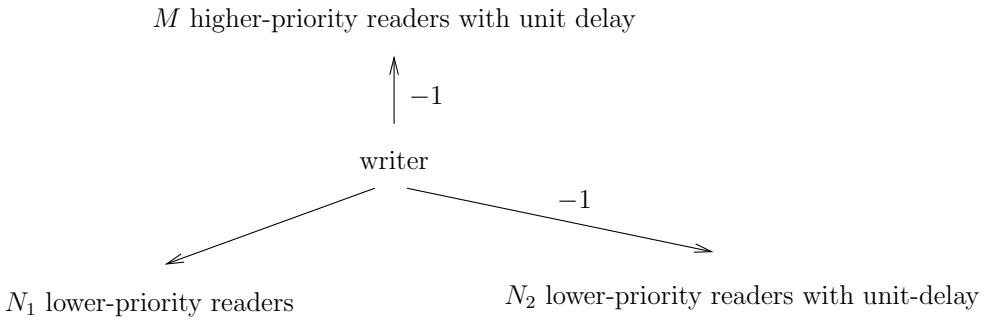



**Figure 7.16:** Applicability of the DBP protocol.

there is no unit-delay, while in the rest $N_2 = N - N_1$ readers there is a unit-delay. DBP can be applied to general (i.e., multi-writer) task graphs as we show in Section 7.5.2.

Apart from being semantics-preserving, DBP also allows to reduce the memory requirements with respect to the simple method presented in Section 7.4.5. In particular, DBP requires $N + 2$ (single) buffers, assuming $M \neq 0$ and $N_2 \neq 0$. If $M = N_2 = 0$ (i.e., there are no readers linked with a unit-delay) then DBP requires $N + 1 = N_1 + 1$ buffers. This is to be compared, for example, to $2N$ buffers required when using the method of Section 7.4.5. Buffer requirements are presented in detail in Section 7.8.

### 7.5.1 The Dynamic Buffering Protocol

DBP is shown in Figure 7.17. The figure shows the protocol in the case where $M \neq 0$ or $N_2 \neq 0$, that is, the case where there are links with unit-delay. If $M = N_2 = 0$ then the protocol is actually simpler: the pointer `previous` is not needed and instead of $N + 2 = N_1 + 2$, only $N_1 + 1$ buffers are needed.

The operation of DBP is as follows. The writer $\tau_w$ maintains all buffers and pointers except the pointers of the higher-priority readers `P[i]`. The `current` pointer points to the position that the writer last wrote to. The `previous` pointer points to the position that the writer wrote to before that. `R[i]` points to the position that $\tau_i$ must read from.

The key point is that when the writer is released, a *free* position in the buffer array must be found, and this is where the writer must write to. By free we mean a position which is not currently in use by any reader, as defined by the predicate `free(j)`. Finding a free $j \in [1..N + 2]$ amounts to finding some $j$ which is different from `previous` (because `B[previous]` may be used by the higher-priority reader or a possible lower-priority with unit-delay reader may need to copy its value) and also different from all `R[i]` (because `B[R[i]]` is used, or will be used, by the lower-priority reader $\tau_i$). Notice that such a $j$ always exists, by the *pigeon-hole principle*: there are $N + 2$ possible values for $j$ and up to $N + 1$ possible values for `previous` and all `R[i]`.

Finding a free position is done in the second instruction executed upon the release of the writer. The first instruction updates the `previous` pointer. This pointer is copied by each higher-priority reader $\tau_i'$, when released, into its local variable `P[i]`. $\tau_i'$ then reads from `B[P[i]]`.

When a lower-priority reader $\tau_i$ is released, we have two cases: (i) either $\tau_i$ is one of the $N_1$ readers that are linked without a unit-delay, or (ii) $\tau_i$ is one of the $N_2$ readers that are linked with unit-delay. In case (i) $\tau_i$ needs the last value written by the writer. In case (ii) $\tau_i$ needs the previous value. Pointer `R[i]` is set to the needed value. Besides this pointer assignment the rest of the procedure remains the same for both kinds of lower-priority readers. While executing, $\tau_i$ reads from `B[R[i]]`. When $\tau_i$ finishes execution, `R[i]` is set to `null`. This is done for optimization purposes, so that buffers can be re-used as early as possible. Notice that even if this operation is removed, DBP will still be correct and it will use at most $N + 2$ buffers. However, DBP will be sub-optimal, in the sense that the buffer pointed to by `R[i]` will not be freed until the next release of $\tau_i$. With the above operation present, the buffer is freed earlier, namely, when the current release of $\tau_i$ finishes.

Notice that DBP also relies on the fact that no more than one instance of every task can be active at any point in time, which follows from the schedulability assumption. In more detail, there can be no more than one pointer per lower-priority task, which allow us to use the *pigeon hole principle* as we did before.

Like the protocols discussed in the previous section, DBP also specifies release actions for both writer and reader tasks. In case of simultaneous release of more than one tasks, we require that *the release actions of the writer task are performed before the release actions of the simultaneously released readers*. The actions of the readers can be performed in any order.

Another major contribution of the above algorithm is that it does not take into account any

Communication: $\tau_w \rightarrow \tau_i$, for $i = 1, ..., N_1$, $\tau_w \stackrel{-1}{\rightarrow} \tau_i$, for $i = N_1 + 1, ..., N_1 + N_2$, and $\tau_w \stackrel{-1}{\rightarrow} \tau'_i$, for $i = 1, ..., M$. Let $N = N_1 + N_2$.
Task $\tau_w$ maintains a buffer array `B[1..`$N$`+2]`, one pointer array `R[1..`$N$`]` and two pointers `current` and `previous`.
Each task $\tau'_i$, for $i = 1, ..., M$, maintains a local pointer `P[i]`.
All pointers are integers in `[1..`$N$`+2]`. A pointer can also be `null`.
Initially, `current = previous = 1`, all `R[i]` and `P[i]` are set to `null`, and all buffer elements are set to $y_0^i$.
During execution:
*Writer*:

- When $\tau_w$ is released:
  `previous := current;`
  `current :=` some `j`$\in$`[1..`$N$`+2]` such that `free(j)`, where
  `free(j)` $\equiv$ `(previous`$\neq$`j` $\wedge$ $\forall$`i`$\in$`[1..`$N$`].R[i]`$\neq$`j).`
- While $\tau_w$ executes it writes to `B[current]`.

*Lower-priority reader* $\tau_i$:

- When $\tau_i$ is released:
  `if i`$\in$`[1..`$N_1$`] then R[i] := current` (link $\tau_w \rightarrow \tau_i$)
  `else R[i] := previous` (link $\tau_w \stackrel{-1}{\rightarrow} \tau_i$)
- While $\tau_i$ executes: reads from `B[R[i]]`.
- When $\tau_i$ finishes: `R[i] := null`.

*Higher-priority reader* $\tau'_i$:

- When $\tau'_i$ is released: `P[i] := previous`.
- While $\tau'_i$ executes: reads from `B[P[i]]`.

Applicable under static-priority scheduling when

$$\forall i = 1, ..., N \; . \; p_i < p_w. \;\; \texttt{and} \;\; \forall i = 1, ..., M \; . \; p'_i > p_w.$$

Applicable under EDF scheduling when

$$\forall i = 1, ..., N \; . \; d_i > d_w \;\; \texttt{and} \;\; \forall i = 1, ..., M \; . \; d'_i < d_w.$$

In case of simultaneous release of writer and readers: the action for the writer is executed first. The actions for the readers can be executed in any order.

**Figure 7.17:** The protocol DBP.

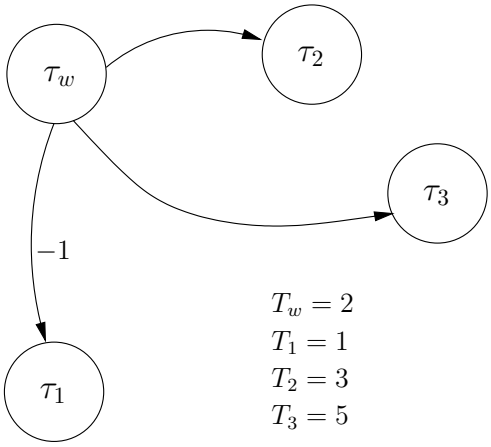



**Figure 7.18:** A task graph with one writer and three readers.

possible initial offset of the concerning tasks. This is important mostly for the multi-periodic application of the algorithm, as we'll study in Section 7.5.3, since methods used up to now, exclude the use of the initial offset.

### An example

To illustrate how DBP works, we provide an example. Consider the task graph shown in Figure 7.18. There are four tasks: one writer $\tau_w$ with period $T_w = 2$, one higher-priority reader $\tau_1$ with period $T_1 = 1$ and two lower-priority readers $\tau_2$ and $\tau_3$ with periods $T_2 = 3$ and $T_3 = 5$ respectively. This means that for the one writer of this task graph $N = 2$, where $N$ is the number of lower priority readers. Moreover there is one task with higher priority. Suppose the priorities of the tasks follow the rate-monotonic assignment policy (note that DBP does not require this, as it can work with any priority assignment):

$$Prio_1 > Prio_w > Prio_2 > Prio_3.$$

According to the algorithm, the writer will maintain a buffer array `B`, a pointer array `R` of size 2, and two pointers `current` and `previous`. Also, $\tau_1$ maintains a local pointer `P`. Note that, since $N = 2$, `B` cannot grow larger than 4 buffers. The initial values are `current=previous=1` and `R[2]=R[3]=P[1]=null`.

A sample execution of DBP is shown in Figures 7.19 and 7.20. Figure 7.20 shows the values of the pointers during execution. Figure 7.19 shows the release, begin of execution and end of execution events for each task. Task $\tau_1$ is released at times $0, 1, 2, 3, 4, 5$, task $\tau_w$ is released at times $0, 2, 4$, and so on. We use the notation $\tau_1, \tau_1', ...$ to denote different instances of the same task. Notice that a task instance may be "split" because of preemption: this is, for instance, the case of $\tau_2$ which is split between the first and second cycle. The heights of the task "boxes" in the figure denote the relative priorities of the tasks.

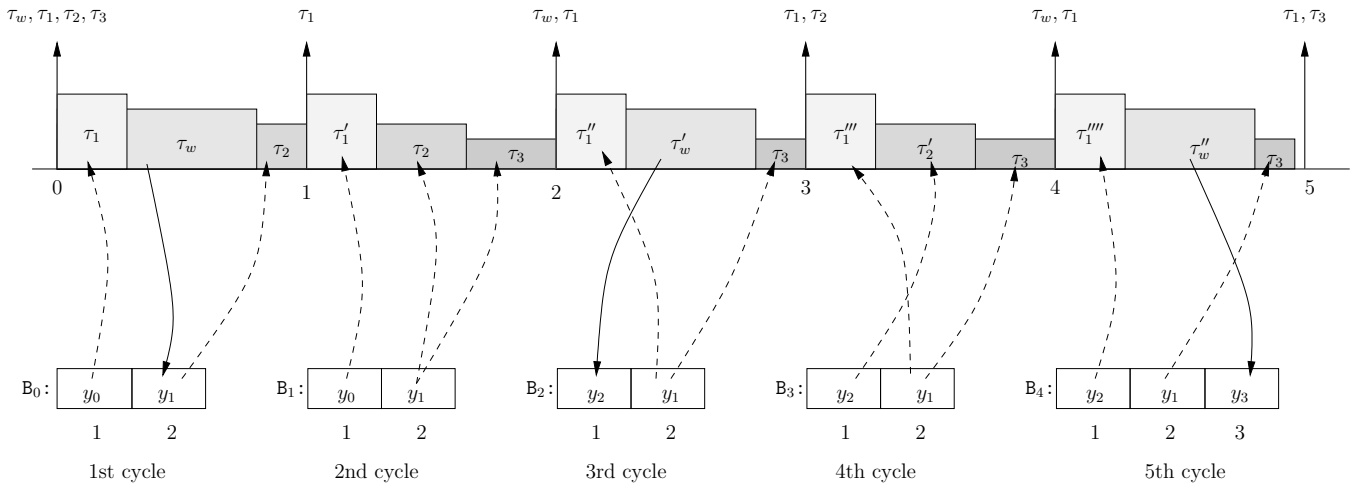



**Figure 7.19:** The execution of the tasks.

| | init | 0 | 1 | 2 | 3 | 4 | 5 |
|---|---|---|---|---|---|---|---|
| `current` | 1 | 2 | 2 | 1 | 1 | 3 | 3 |
| `previous` | 1 | 1 | 1 | 2 | 2 | 2 | 2 |
| `P[1]` | null | 1 | 1 | 2 | 2 | 1 | 1 |
| `R[2]` | null | 2 | 2 | null | 1 | null | null |
| `R[3]` | null | 2 | 2 | 2 | 2 | 2 | 3 |

**Figure 7.20:** The values of the DBP pointers during execution.

Figure 7.19 also shows exactly where each task reads from and writes to at any given time (dashed and solid arrows respectively). The "boxes" at the bottom of the figure correspond to the buffer array `B` and the values stored in each buffer: $y_0$ is the initial (default) value, $y_1$ is the value written by the first instance of $\tau_w$, and so on. Notice that `B` grows to 3 buffers in this example.

It can be verified that the synchronous semantics are preserved. For example, the first instance of reader 2, $\tau_2$, reads the value produced by the first instance of the writer, which was released at the same time. The third instance of reader 1, $\tau_1''$, reads the same value: this is because a unit delay is present in this case. It is worth noting that the unique instance of reader 3 shown in the figure, although it is preempted multiple times, consistently reads the correct value, namely $y_1$. The fact that this instance has not terminated execution when the writer is released at time $4$ is what triggers the allocation of a new buffer `B[3]`.

### Specializations

It can be easily shown that the three protocols presented in Section 7.4 are specializations of DBP. First, consider the low-to-high protocol (Figure 7.5). It can be obtained by using DBP with

$N = 0$ and $M = 1$. Then, DBP uses two buffers `B[1..2]` and three pointers: `previous, current, P[1]`. In fact, `previous` is redundant since it always points to the buffer not pointed to by `current`. Thus, `current` corresponds to the pointer `current` of Figure 7.5 and `P[1]` corresponds to the pointer `previous` of Figure 7.5 (the latter is local to the reader).

Next, consider the high-to-low protocol (Figure 7.7). It can be obtained by using DBP with $N_2 = M = 0$ and $N_1 = 1$. As we said above, in this case only $N_1 + 1 = 2$ buffers are needed and the pointer `previous` is useless. Thus, DBP uses two pointers `current, R[1]`: they correspond to pointers `next` and `current` of Figure 7.7, respectively.

Finally, consider the high-to-low protocol with unit-delay (Figure 7.9). It can be obtained by using DBP with $N_1 = M = 0$ and $N_2 = 1$. Then, DBP uses three buffers `B[1..3]` and three pointers `previous, current, R[1]`: they correspond to pointers `previous, current, reading` of Figure 7.9, respectively.

### 7.5.2 Application of DBP to general task graphs

Applying DBP to a general task graph is easy: we consider each writer task in the graph and apply DBP to this writer and all its readers. As an example, let us consider again the task graph shown in Figure 7.15. There are three writers in this graph, namely, $\tau_1$, $\tau_3$ and $\tau_4$. We assume, for each writer, that it communicates the same data to all its readers. Then, the buffer requirements are as follows:

- $\tau_1$ has only one lower-priority reader without unit-delay. That is, we are in the case $M = N_2 = 0$ and $N_1 = 1$. As said above, in this case DBP specializes to the high-to-low protocol, which requires one double buffer.

- $\tau_3$ has two higher-priority readers $\tau_1$ and $\tau_2$ (with unit-delay), one lower-priority reader $\tau_4$ without unit-delay and one lower-priority reader $\tau_5$ with unit-delay. That is, we are in the case $N_1 = N_2 = 1$ and $M = 2$. We apply DBP and we need $N + 2 = 4$ buffers.

- $\tau_4$ has two higher-priority readers. That is, we are in the case $N = 0$ and $M = 2$. We apply DBP and we need 2 buffers.

Thus, in total, we have $8$ single buffers. This is to be compared to $11$ single buffers needed using the method described in Section 7.4.5.

In the case that we have more than one writer, as we said earlier, there is distinctive application of the DBP protocol with new memory space and pointers. Therefore, we do not need to take care about the execution of assignments implied by two different instances of DBP (we could impose for example a partial order for the readers and the writers depending on the priority of the writers of the corresponding DBP). This means that the assumption we expressed earlier, i.e. when a reader and a writer arrive simultaneously, we give priority to the actions of the writer, is sufficient in the general case, we just described.

### 7.5.3 Application of DBP to tasks with known arrival pattern

We have seen that the three protocols, described in Section 7.4, are correct for any arrival pattern. The same holds for DBP as well. In this section we examine the application of DBP to applications where the triggering pattern of tasks is known, for instance, *multi-periodic* applications where each task is periodic with a known period. We will see that in this case DBP can be optimized, taking advantage of the additional knowledge on the arrival patterns. In this section we discuss a time optimization, which aims at turning DBP into a *static* rather than dynamic protocol. In Section 7.8 we discuss memory optimizations.

DBP, being a dynamic protocol in general, implies an overhead, namely, searching for a free buffer to use every time the writer is released. To this overhead must be added the time to allocate a buffer on-the-fly, or the "waste" of memory in case buffers are pre-allocated. We can do better if we know the arrival pattern of the tasks, by providing a *static* schedule. This can be done by simulating the DBP with the, *a priori* known, release times of the reader(s) and writer tasks. During simulation we keep track where the writer stores data on every execution and from where the reader(s) read from, on every execution as well. Those positions are instructed by DBP and more precisely, for the writer it is the `current` pointer and for the reader task $\tau_i$ it is the pointer `R[i]`.

This simulation must be done until the *hyper-period* of the tasks. In the case where the tasks are multi-periodic, this hyper-period $T_h$ is the least common multiplier of all periods. If tasks are not multi-periodic then we will have to find a point where the arrival pattern starts repeating. If that point doesn't exist either, we will have to keep track for the entire possible execution of the system. This last one, of course, may be very inefficient in case the execution is not short, causing the static schedule to be enormous in terms of memory (to remember where the readers and writer read from and writes to).

## 7.6 Periods in consecutive powers of two

We further investigate special cases of applications for which the periods are powers of two. We also consider one additional factor, whether we have atomic reads and writes or not. If data transfer is instantaneous then we can further optimize the system since the scheduling constraints allow earlier release of buffer space.

### 7.6.1 Non-atomic case

Consider $n$ tasks, $\tau_1, ..., \tau_n$, such that task $\tau_i$ has period $T_i = 2^{i-1}$ and a priority which is in inverse order of period. Suppose each task sends data to all lower-priority tasks. For three tasks of periods $T_1 = 1$, $T_2 = 2$ and $T_3 = 4$, this gives the situation shown in Figure 7.21.

We assume that for simultaneous occurrences, the higher-priority task takes precedence and thus transmits its data to the co-incidentally occurring lower-priority task. Thus the first emission of $\tau_1$ (`r1a`) is required by both $\tau_2$ and $\tau_3$. Furthermore, the data is required to persist until the end of the period of $\tau_3$, i.e., until `r3a'`. However, the same emission is required by $\tau_2$ and this

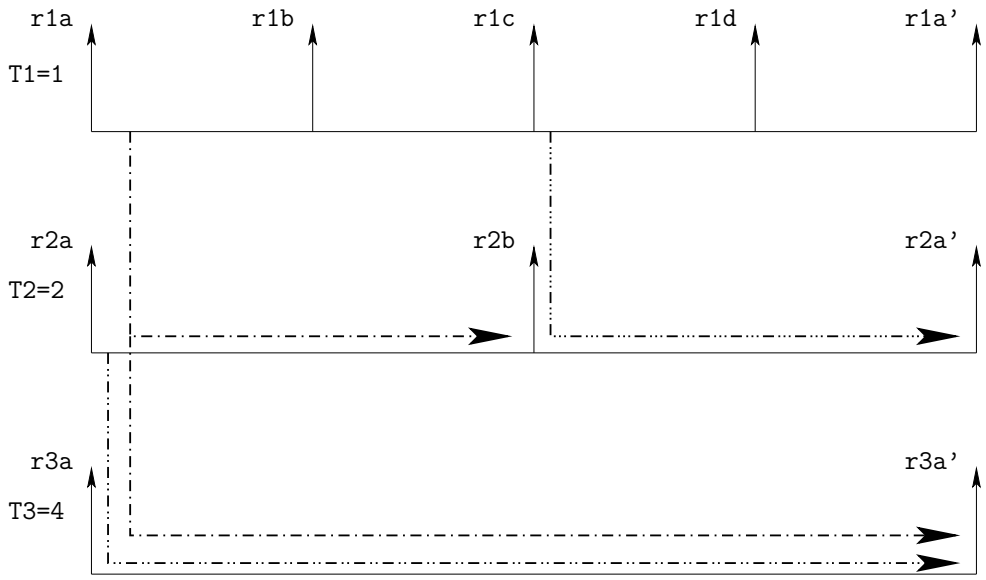



**Figure 7.21:** Multiperiodic events with consecutive periods in powers of 2

buffer can be shared with $\tau_3$. Thus we can implement this system using three single buffers (one for `r1a` to `r2a` and `r3a`, one for `r2a` to `r3a` and one for `r1c` to `r2b`). Our original scheme would require three double buffers.

In fact, it is immediately apparent that for the highest-priority task out of $n$ tasks we require $n-1$ single buffers. Thus, for each writer task $i$, we require $n-i$ single buffers, one for each of the (lower-priority) reader tasks. When every task is a writer, this gives a total of $n(n-1)/2$ single buffers. If we used the general scheme of Subsection 7.4 we would require $n(n-1)/2$ double buffers, that is, double the memory space.

To implement this scheme in practice requires a buffer indexing mechanism. Generating the indexing pattern implied by Figure 7.21 is quite simple. We must first observe that each alternate emission of the top-level task is unused by lower-priority tasks. Given this, for one writer and $n$ reader tasks, for emission $i = \{0, 2, 4, \ldots\}$ the writer utilizes buffer:

$$B(i) = \min\left\{j : 0 \leq j \leq n-1, i \bmod 2^{n-j} = 0\right\} \tag{7.1}$$

The writer at time $i$ writes to buffer $B(i)$, any reader occurring at time $i$ reads from buffer $B(i)$. For example, for $n = 3$ we need 3 buffers indexed $0, 1, 2$ and used according to the pattern $B(0) = 0$, $B(2) = 2$, $B(4) = 1$, $B(6) = 2$, $B(8) = 0$, and so on. A potentially useful value is the residence time of the data in the buffer which can be computed as: $R(i) = 2^{n-B(i)}$.

### 7.6.2 Atomic case

In Section 7.6 we made no assumptions about the atomicity of data transfers and we were constrained to allow buffers to be occupied for the entire period of a given task. However, we know

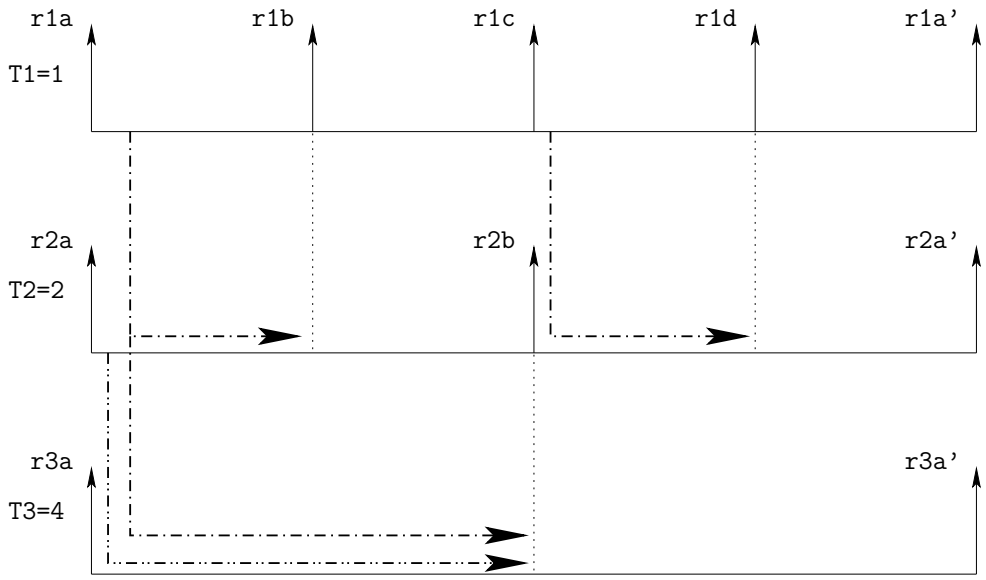



**Figure 7.22:** Multiperiodic events with consecutive periods in powers of 2, atomic reads

from the fixed-priority scheduling constraints that a low-priority task $\tau_j$ must begin execution before the mid-point of its period, i.e., $2^{j-2}$. If it has not managed to do so, this means that the cumulative execution time of higher-priority tasks is at least half the period of $\tau_j$ and, since all these higher-priority tasks will also be executed at the second half of $\tau_j$'s period, $\tau_j$ will never get to execute. This contradicts our assumption that the system is schedulable.

Now, we assume that tasks sample their data at the start of execution so if we can arrange for reading of the data to be completed prior to the period mid-point then the related buffer becomes free before the next emissions of higher-priority tasks. If we have a fixed bound on communications we can simply add this time to the execution time for each higher-priority task and the scheduler can then guarantee atomicity in the sense described here. Alternatively, we could rely upon the operating system to provide atomic data transfer. In any case, we can make further savings in the number of required buffers.

Consider Figure 7.22 which illustrates the situation for three tasks and atomic reads. We can now implement communications from $\tau_1$ to tasks $\tau_2$ and $\tau_3$ using only a single buffer because we can use the same buffer for `r1a` to `r2a` and for `r1c` to `r2b`. In fact, for writing task $\tau_i$, we need $\lfloor (n-i)/2 \rfloor$ buffers. The reason for the divide by two is that since only every second emission from task $\tau_i$ is needed we do not need a new buffer each time we add a new higher-priority task. To see this consider the two processor case in Figure 7.22 ($\tau_2$ and $\tau_3$) which needs a single buffer (from `r2a` to `r3a`) and compare with the three processor case in the same figure. Since `r2b` is ignored, no further buffers are required. For $n$ tasks, with $n$ even we need $(n/2)^2$ buffers, and for $n$ odd we need $(n/2)^2 + (n/2)$ buffers in total.

The buffer indexing function for this situation is simply $B(i)/2$ and the data residence time $R(i)/2$.

Non-consecutive powers of two do not pose any serious problems since these are simply subsets of the current analysis. The only complication is that the buffer indexing may require support in the form of index tables rather than analytical formulae as above. More general harmonic cases can also be treated using the methods here, for example periods of $T_1 = 1, T_2 = 2$ and $T_3 = 6$ are almost the same as for the $1 - 2 - 4$ case apart from the duplicate emissions from $\tau_1$ to $\tau_2$.

## 7.7 Proof of correctness

In this section we formally prove that the protocols proposed in Sections 7.4 and 7.5 are correct, that is, they preserve the ideal semantics. We will use two different proof techniques. For the one-writer/one-reader protocols of Section 7.4 we reduce correctness to a problem of *model-checking* on a *finite-state* model, where automatic verification techniques can be applied [QS81, CGP00]. For the general protocol of Section 7.5 we provide a "manual" proof. Although the latter establishes the correctness of the special protocols as well, we believe that the model-checking proof technique is still worth presenting, because it can serve to establish correctness of other similar protocols in an automatic way. Moreover, from a historical point of view, the one-writer to one-reader protocols, have been founded earlier and the model-checking proof was used in the first place.

### 7.7.1 Proof of correctness using model-checking

In order to use model-checking, we must justify why finite-state models suffice. We do this in a series of steps.

The first step is to prove correctness of each protocol for a single writer and a single reader. Obviously, the protocols must function correctly for an arbitrary number of tasks. However, we do not want to model all these tasks, since this would yield a model with an unbounded number of tasks, where model-checking is not directly applicable. To avoid this, we employ the following argument. We claim that proving correctness of a given protocol *only for two tasks, one writer and one reader*, is sufficient, *provided the effect of other tasks on these two tasks is taken into account*.

But what is the "effect of other tasks"? In the case where a protocol of Section 7.4 is used only for a single writer/reader link, the buffers are not shared with the other tasks. Therefore, the only way the other tasks influence the writer/reader pair in question is by preemption. We will show how to model this influence, although we are not going to model preemption explicitly.

Having eliminated the problem of infinite number of tasks, we still have the problem of *data types*. Our buffering protocols are able to convey any data type. However, in order to use model-checking directly, variables must take values in a finite domain. To solve this problem we use the technique of uninterpreted functions [BD94], which allows to replace the unknown data type with a fixed number of distinct values. This number is the maximum number $m$ of distinct values that can be present in the system at the same time. To implement this idea, we replace the data type with a $n$-vector of booleans, such that $2^n \geq m$.

The general architecture of the model we shall use for model-checking is shown in Figure 7.23. The model has four components. An *event generator* component which produces the events of the tasks. A component modeling the *ideal semantics*. A component modeling the behavior of the *buffering protocol*. A *preservation monitor* component which compares the two behaviors and checks whether they are "equivalent" (where the notion of equivalence is to be defined).

The above described approach, using the model based paradigm, is interesting because it provides means to verify the preservation of semantics in a larger extend. One can use this scheme to generate sequences of inputs and compare the results of the protocol in question with respect to an automaton representing some "ideal" behavior, or some behavior under test.

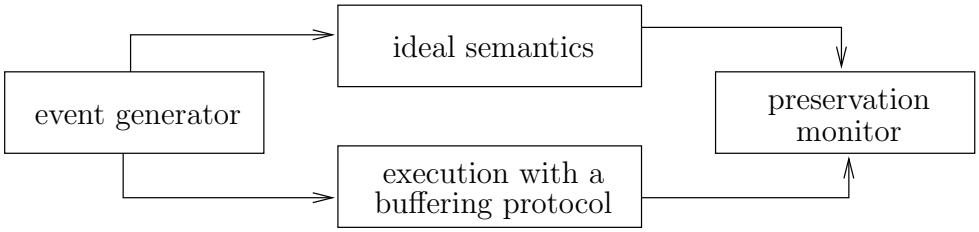



**Figure 7.23:** Architecture of the model used in model-checking.

### Event generator model

A task $\tau_i$ is modeled by three events, $r_i$, $b_i$ and $e_i$, corresponding to the release, beginning of execution and end of execution of (an instance of) the task, respectively. In the ideal semantics, these three events occur simultaneously. In the real implementation, these events follow a *cyclic* order

$$r_i \rightarrow b_i \rightarrow e_i \rightarrow r_i \rightarrow b_i \rightarrow e_i \rightarrow \cdots,$$

which corresponds to two facts. First, that each task instance is first released, then starts execution and finally ends execution. Second, that when a new instance is released the previous instance has finished, which is our schedulability assumption. Notice that although preemption may occur between $b_i$ and $e_i$, they are not modeled explicitly.

The scheduling policy is captured by placing restrictions on the possible *interleavings* of the above events. Let us first show how to model static-priority scheduling. Let $\tau_1$ be the high-priority task and $\tau_2$ be the low-priority task. Then, we know that neither $b_2$ nor $e_2$ can occur between $r_1$ and $e_1$. Indeed, $\tau_2$ cannot start before $\tau_1$ finishes. Also, if $\tau_2$ has already started when $r_1$ occurs then it is preempted, thus, will not finish before $\tau_1$ finishes. These ordering restrictions can be modeled using the finite-state automaton shown in Figure 7.24.[3]

The automaton has five states. The state labeled "false" corresponds to the violation of the static-priority scheduling assumption. In other words, the legal orderings of the events $r_i, b_i, e_i$

[3] For simplicity, the automaton shown in the figure assumes that no two events can occur simultaneously. This assumption can be lifted but it results in a more complicated automaton which is not shown.

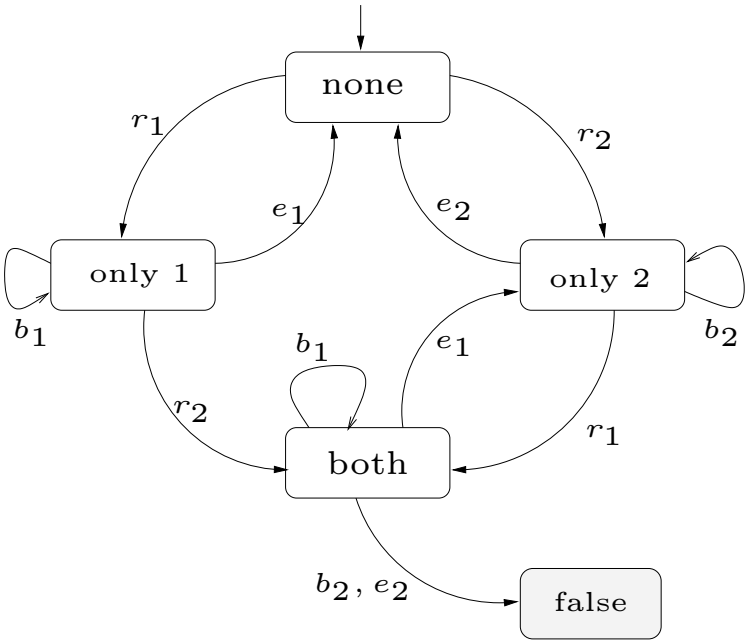



**Figure 7.24:** Assumptions modeling static-priority scheduling: $p_1 > p_2$.

are those orderings where the "false" state is not reached. For example, $r_2 r_1 b_1 e_1 b_2 e_2$ is legal, but $r_2 r_1 b_2 e_2$ is not. The other four states correspond to the cases where no task, only one task, or both tasks have been released.

EDF scheduling can be modeled in a similar way. Let $\tau_1$ be the task with the smaller deadline and $\tau_2$ be the task with the larger deadline. Figure 7.25 shows an automaton modeling this case. Again, when state "false" is reached the EDF scheduling assumption is violated. Note that the restrictions imposed by the automaton of Figure 7.25 are weaker than those imposed by the automaton of Figure 7.24. For example, the sequence $r_2 r_1 b_2 e_2 b_1 e_1$ is accepted by the former automaton but not by the latter. This sequence corresponds to the case where the absolute deadline of $\tau_1$ is greater than the one of $\tau_2$, thus, $\tau_2$ is not preempted.

### Ideal semantics model

The other three models are described in the synchronous language LUSTRE. In fact, this is the language we used for model-checking. In LUSTRE, as we have seen, variables denote infinite sequences of values, the *flows*. The ideal semantics for the case $\tau_1 \rightarrow \tau_2$ can be described in LUSTRE as shown in Figure 7.26. A similar model can be built for the case $\tau_1 \overset{-1}{\rightarrow} \tau_2$.

The boolean flows `r1` and `r2` model the events $r_1$ and $r_2$. That is, event $r_1$ occurs when and only when `r1` is "true", and similarly for $r_2$. These flows are generated by the event generator model presented previously.

The flow `val` models the values written by the writer task. It is an $n$-vector of booleans, according to the abstraction technique explained previously. This flow is also generated externally,

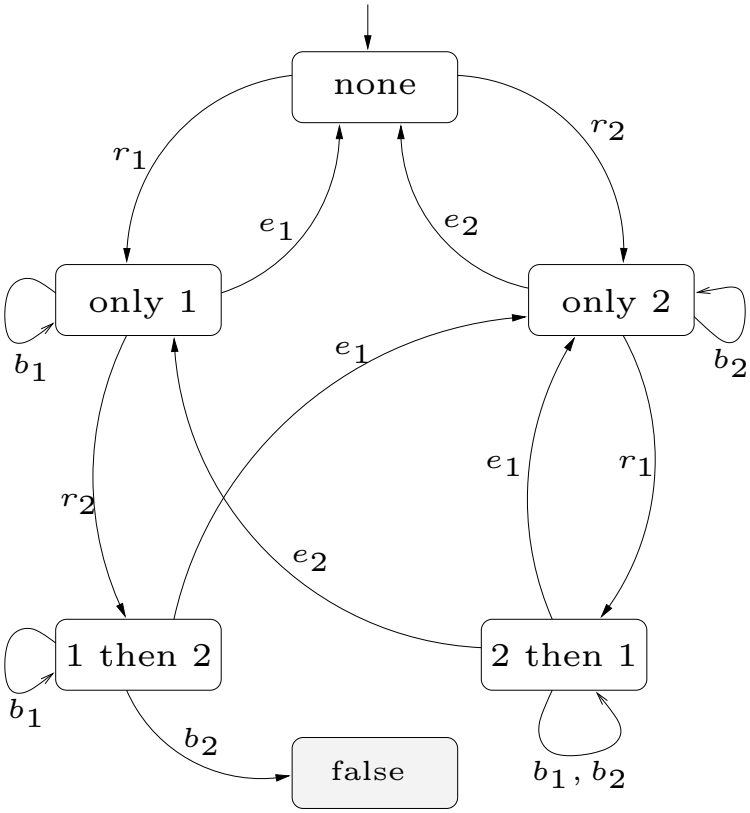



**Figure 7.25:** Assumptions modeling EDF scheduling: $d_1 < d_2$.

```
ideal1 = if r1 then val else (init -> pre ideal1);
ideal2 = if r2 then ideal1 else (init -> pre ideal2);
```

**Figure 7.26:** The ideal semantics described in LUSTRE: the case $\tau_1 \rightarrow \tau_2$.

```
node Low_to_High(r1, r2, e1: bool; w_val: bool^n)
returns (r_val: bool^n);
var buff_0, buff_1: bool^n;
    curr, prev: bool;
let

  curr   = false -> if r1 then not pre curr else pre curr;
  prev   = false -> if r2 then not curr     else pre prev;

  buff_0 = if e1 and curr     then w_val else (init -> pre buff_0);
  buff_1 = if e1 and not curr then w_val else (init -> pre buff_1);

  r_val  = if prev then buff_0 else buff_1;
tel
```

**Figure 7.27:** The low-to-high protocol described in LUSTRE.

in a totally non-deterministic manner (i.e., all possible values are explored in model-checking).

The flow `ideal1` models the output of the writer task. The output is initialized to value `init`, also provided externally (this is the LUSTRE expression `init -> ...`). The output is updated to `val` every time $r_1$ occurs. Otherwise it keeps its previous value (**pre** `ideal1`).

The flow `ideal2` models the input of the reader task. The input is initialized to `init` and is updated to the output of the writer every time $r_2$ occurs. Otherwise it keeps its previous value.

### Buffering protocol model

The buffering protocol is also modeled in LUSTRE. Figure 7.27 shows the model for the low-to-high protocol. Similar models are built for the other protocols.

The model is a LUSTRE *node*, similar to a C function. The node takes as inputs boolean flows `r1, r2, e1` (corresponding to events $r_1, r_2, e_1$) and n-vector boolean flow `w_val` (corresponding to the output of the writer) and returns the flow `r_val` (corresponding to the input of the reader). The flows `buff_0, buff_1, curr, prev` are internal variables corresponding to the double buffer and boolean pointers manipulated by the protocol (see Figure 7.5).

Although event $e_1$ is not a trigger of the low-to-high protocol, it is used in the modeling in order to update the double buffer: the latter is updated when $e_1$ occurs, i.e., when the writer task finishes.

### Preservation monitor model

The preservation monitor is also modeled in LUSTRE, as shown in Figure 7.28. The monitor verifies whether the input of the reader task in the ideal semantics, `ideal2`, is always equal to the input of the reader task as this is produced by the `Low_to_High` node (`vecteq` is a function that checks equality of vectors). The only subtlety is that this check is performed only

```
verif_period = until(b2, e2);
prop = if verif_period
    then vecteq(ideal2, Low_to_High(r1, r2, e1, ideal1))
    else true;
```

**Figure 7.28:** Model checking described in LUSTRE.

at certain moments in time, and in particular during the interval from the beginning until the end of execution of the reader task. This interval is captured by the boolean flow `verif_period`. Indeed, outside this interval the values of the reader input in the ideal and real semantics are generally different.

### Verification using LESAR

We performed model-checking of the models described above using LESAR, the model-checker associated to the LUSTRE tool-suite [RHR91]. For the first two cases, the verification of the hi-to-low and low-to-hi protocols, the model checker, replied with `TRUE PROPERTY`, i.e., that the protocol is always equal to the ideal semantics, in less than half minute time and using almost $100.000$ states for this computation. On the other hand, for the hi-to-low with unit-delay, the computation time was more than $4$ minutes, the use of states exceeded the $520.000$ and the result was also `TRUE PROPERTY`.

The above computation time and state space, is for the verification of our protocols given that we assert static-priority scheduling, i.e., that the generated releases, starts and ends of the tasks have fixed priorities, as seen in Figure 7.24. However, using the EDF scheduling policy there is a bigger "freedom" in the generation of those events, resulting to larger computation time and state space. Indeed, for the verification of the hi-to-low protocol with a unit-delay, with an EDF scheduler, the computation time is more than $40$ minutes and the state space reaches the $1.700.000$ states, to prove the `TRUE PROPERTY` as before.

### Proof of correctness of the optimized buffering schemes, harmonic case

A manual proof of this should be possible but for simplicity we merely adapt our model-checking proof of the general case by extending to periodic systems with appropriate periods. Figure 7.29 shows the buffering scheme for this model.

The `pat` value contains the computed buffer index pattern from Equation 7.1. The nodes `Iinc`, `Ieq` etc. implement integer arithmetic in boolean arrays. Thus the `cnt` variable counts out the source task occurrences and is the correct width (`mdivn`) such that it resets to zero when the occurrences match. The value `mdivn` is thus $\log_2(T_j/T_i)$ so for example for communication from Task $\tau_1$ to $\tau_3$ `mdivn` is 2 and only every fourth emission is read (controlled by the `bitfrom` flag). The `idx` value is the index into the `pat` array and is incremented on each significant source event. The actual buffer index `bufi` is read out of the `pat` array and used to select which buffer (`bufsin`) to use. The `Bget` and `Bset` routines index an array of buffers

```
node singlebuf3_p21b(const t, t0, t2, b, b2, mdivn: int;
                     fromev, toev, fromact, tobeg: B;
                     fromval: T; bufsin: T^b)
returns (toval: T; bitfrom: B; bufsout: T^b; idx: B^t;
         cnt: B^mdivn; bufi, bufo: B^b2);
var c1, c2: B; pat: B^b2^t2; cntnext: B^mdivn;
        idxnext: B^t; even: T;
let
  pat = indxs(t, t2, b2);
  cntnext, c2 = Iinc(mdivn, pre cnt);
  cnt = Iminusone(mdivn) ->
           if ist(fromev) then cntnext else pre cnt;
  bitfrom = Ieq(mdivn, cnt, Izero(mdivn));
  idxnext, c1 = Iinc0(t0, t, pre idx);
  idx = Iminusone(t) ->
        if ist(fromev) and ist(bitfrom) then idxnext
                                        else pre idx;
  bufi = Bget(t2, t, b2, pat, idx);
  even = if ist(fromact) and ist(bitfrom)
         then fromval else (init -> pre even);
  bufsout = BsetT(b, b2, bufsin, bufi, even);
  bufo = if ist(toev) then bufi else fls^b2 -> pre bufo;
  toval = if ist(tobeg)
          then if ist(fromact) and ist(bitfrom)
               then fromval
               else BgetT(b, b2, bufsin, bufo)
          else (init -> pre toval);
tel
```

**Figure 7.29:** Single buffer scheme for periods in powers of 2

using boolean arrays as indices. The appropriate buffer is then written to (`bufsout`) and the buffer index is latched for the receiving process in `bufo`. The correct value is then read out of the buffer when the receiving task begins execution (`tobeg`).

This model-checks correctly for $n = 3$ but is too big to verify for $T_4 = 8$. Partial verification is possible and all subsets are true for $n = 4$. Since there are no irregularities in this construction (each case is built by extending the previous one in a regular way) we can conclude that the scheme is correct for all $n$. For systems with contiguous priorities in powers of 2 we can thus implement our buffering mechanism using the same number of single buffers as we would need double buffers in the general case.

#### Buffering for consecutive powers of two, atomic case

Using the same code as in Figure 7.29, but replacing the buffer pattern `pat` with the buffer pattern implied by $B(i)/2$ and updating the priority conditions to reflect the new scheduling constraints model-checks correctly for the $n = 3$ case. Again, we are only able to partially verify the mechanism for four tasks.

Model-checking can also be used to prove correctness of the optimized versions, however, this can be done *a priori* only for a given, rather than arbitrary, set of tasks. We have followed this approach and modeled the buffering schemes for the harmonic multi-periodic case (Sections 7.6 and 7.6.2). In both cases, we have managed to model-check completely only systems of $n = 3$ tasks. The model gets too large for LESAR to handle for $n = 4$. We did manage, however, to partially verify the $n = 4$ case, selecting various subsets of the model and proving them correct. Since there are no irregularities in these schemes (each case is built by extending the previous one in a regular way) we can expect the scheme to be correct for all $n$.

### 7.7.2 Proof of correctness of the dynamic buffering protocol

In this section we will prove the correctness of the protocol DBP (Section 7.5.1). In this case, model-checking is not directly applicable, since we have an arbitrary number of reader tasks. Instead, we provide a "manual" proof.

What we want to prove is semantical preservation, that is, that for any possible arrival pattern and values written by the writer, the values read by the readers in the ideal semantics are equal to the values read by the readers in the implementation, assuming DBP is used. More formally, consider a reader $\tau_i$ and let $t_i$ be the time when an arbitrary instance of $\tau_i$ is released. We denote this instance by $\tau_i^{t_i}$. Let $t_i' \geq t_i$ be the time when $\tau_i^{t_i}$ reads. Let $\tau_w$ be the writer task. For the moment, let us assume that $\tau_w$ is released at least twice before time $t_i$. We relax this assumption later in this section.

Let $t \leq t_i$ be the last time before $t_i$ that an instance of $\tau_w$ was released. We denote this instance by $\tau_w^t$. Let $t_e > t$ be the time that $\tau_w^t$ produces its output and finishes. Let $y(t)$ be the output of $\tau_w^t$. Let $t' < t$ be the last time before $t$ that an instance of the writer $\tau_w$ was released. This instance is denoted $\tau_w^{t'}$. It finishes execution at time $t_e' > t'$. Let $y(t')$ be the output of $\tau_w^{t'}$. Figure 7.30 illustrates the notation defined above. Notice that the order of events shown in the figure is just one of the possible orders.

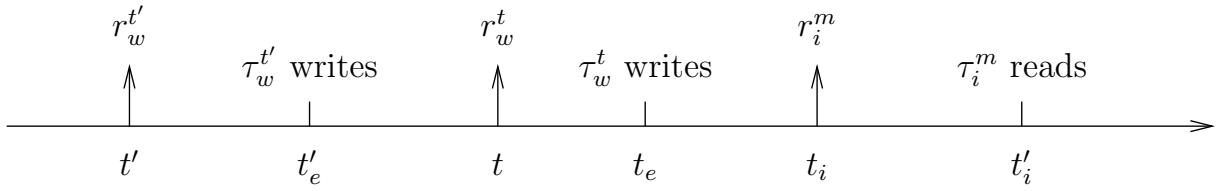



**Figure 7.30:** Illustration used in the proof of DBP.

### Lower-priority reader without unit-delay

Suppose, as a first case, that the reader $\tau_i$ has a lower priority than the writer $\tau_w$ and we have $\tau_w \rightarrow \tau_i$. Let $x(t_i)$ be the value read by $\tau_i^{t_i}$. The ideal semantics states that $x(t_i) = y(t)$. We want to show that this equality holds in the implementation as well.

Let us first handle the case where the writer is never released before time $t_i$. In this case, $y(t)$ is equal to the default output of $\tau_w$. Also, when $\tau_i^{t_i}$ is released, `R[i]` is set to 1, which is the initial value of `current` (Figure 7.17). `R[i]` is not modified in the interval $[t_i, t'_i]$. Thus, at time $t'_i$, $\tau_i$ reads the value stored in buffer `B[1]`. This is the default output of $\tau_w$, since no buffer has been written by the writer yet.

Let us now turn to the case where the writer is released at $t < t_i$. Recall that the writer chooses upon release a "free" position in the buffer array where it will write to (Figure 7.17). Such a free position always exists by the pigeon-hole principle, as already mentioned. Let $j^t$ be the position that $\tau_w^t$ chooses. Let $R^t, p^t$ and $c^t$ be the values of `R`, `previous` and `current` at time $t$, right after the execution of the assignments `previous := current` and `current := ...`. Then, by definition of DBP, the following hold:

$$p^t \neq j^t \text{ and } \forall i \in [1..n].R^t[i] \neq j^t \text{ and } c^t = j^t.$$

$t_e \geq t$ is the time when $\tau_w^t$ writes: let $B^{t_e}$ be the value of `B` after this write operation[4]. Then, since `current` is not modified between $t$ and $t_e$ and $c^t = j^t$, we also have:

$$B^{t_e}[j^t] = y(t).$$

Now consider the reader $\tau_i^{t_i}$. Again, `current` is not modified between $t$ and $t_i$, thus, we have:

$$R^{t_i}[i] = c^t = j^t.$$

$\tau_i^{t_i}$ reads the value

$$B^{t'_i}[R^{t'_i}[i]] = B^{t'_i}[R^{t_i}[i]] = B^{t'_i}[j^t].$$

This is because `R[i]` is not modified between $t_i$ and $t'_i$.

To show that $\tau_i^{t_i}$ read the correct value $y(t)$, we must show that $B^{t'_i}[j^t] = B^{t_e}[j^t]$, that is, that the position $j^t$ is not over-written between $t_e$ and $t'_i$. This is because only the writer can write into `B[`$j^t$`]` and in order to do so it must choose $j^t$ as a free position. Since the writer does not

[4] Notice that in Figure 7.30 we have $t_e < t_i$ but this need not be the case. We could also have $t_e > t_i$.

arrive in the interval $[t_e, t_i]$, it suffices to show that `free(`$j^t$`)` is false in the interval $[t_i, t_i']$. This is because `R[i]` equals $j^t$ in all this interval.

**Lower-priority reader with unit-delay**

Suppose, next, that the reader $\tau_i$ has a lower priority than the writer $\tau_w$ and we have a link with a unit-delay: $\tau_w \xrightarrow{-1} \tau_i$. Again, let $x(t_i)$ be the value read by $\tau_i^{t_i}$. The ideal semantics states that $x(t_i) = y(t')$. We want to show that this equality holds in the implementation as well.

Let us first handle the case where the writer is released not more than once before time $t_i$. In this case, $y(t')$ is equal to the default output of $\tau_w$. Also, when $\tau_i^{t_i}$ is released, `R[i]` is set to 1, which is the value of `previous` at this point. Indeed, either the writer has never been released yet and `previous` is equal to its initial value 1, or the writer has been released once and `previous` is set to the initial value of `current`, which is also 1. `R[i]` is not modified in the interval $[t_i, t_i']$. Thus, at time $t_i'$, $\tau_i$ reads the value stored in buffer `B[1]`. If the writer has not been released before $t_i$ then `B[1]` holds the default output of $\tau_w$. If the writer has been released once before $t_i$ then it has not written to `B[1]`: to do so, it must choose 1 as a free position to assign to `current`, however, 1 is not free because `previous=1`.

Let us now turn to the case where the writer is released twice before $t_i$. Upon arrival of the writer at time $t'$, a free position in the buffer array is chosen to write to: let this position be $j^{t'}$. Let also $R^{t'}, p^{t'}$ and $c^{t'}$ be the values of `R`, `previous` and `current` at time $t'$, right after the execution of the assignments to `previous` and `current`. Then, by definition of DBP, the following hold:

$$p^{t'} \neq j^{t'} \text{ and } \forall i \in [1..n].R^{t'}[i] \neq j^{t'} \text{ and } c^{t'} = j^{t'}.$$

$t_e' \geq t'$ is the time when $\tau_w^{t'}$ writes: let $B^{t_e'}$ be the value of `B` after this write operation. Then, since `current` is not modified between $t'$ and $t_e'$ and $c^{t'} = j^{t'}$, we also have:

$$B^{t_e'}[j^{t'}] = y(t').$$

On the next arrival of the writer at time $t$, new assignments will be made to the pointers `previous` and `current`. Let $j^t$ be the new free position chosen. Let $R^t, p^t$ and $c^t$ be the values of `R`, `previous` and `current` at time $t$, right after the assignments to `previous` and `current`. Then, the following hold (mention also that $current = c^{t'} = j^{t'}$ remains unchanged from time $t_e'$ and until before the assignments at time $t$):

$$p^t = c^{t'} \; and \; p^t \neq j^t \; and \; \forall i \in [1..n].R^t[i] \neq j^t \; and \; c^t = j^t$$

When the reader arrives at time $t_i$, `R[i]` is set to `previous`. `previous` is not modified between $t$ and $t_i$, thus, the value of `R[i]` at time $t_i$ is equal to $p^t$:

$$R^{t_i}[i] = p^t = c^{t'} = j^{t'}.$$

`R[i]` is not modified between $t_i$ and $t_i'$. Thus, $\tau_i^{t_i}$ reads the value

$$B^{t_i'}[R^{t_i'}[i]] = B^{t_i'}[R^{t_i}[i]] = B^{t_i'}[j^{t'}].$$

To show that $\tau_i^{t_i}$ reads the correct value $y(t')$, we must show that $B^{t'_i}[j^{t'}] = B^{t'_e}[j^{t'}]$, that is, that the position $j^{t'}$ is not over-written between $t'_e$ and $t'_i$. This is because only the writer can write into `B[`$j^{t'}$`]` and in order to do so it must choose $j^{t'}$ as a free position. Since the writer does not arrive in the interval $[t'_e, t]$, it suffices to show that `free(`$j^{t'}$`)` is false in the interval $[t, t'_i]$. In the interval $[t, t_i]$, `free(`$j^{t'}$`)` is false because `previous` equals $j^{t'}$. In the interval $[t_i, t'_i]$, `free(`$j^{t'}$`)` is false because `R[i]` equals $j^{t'}$.

**Higher-priority reader (with unit-delay)**

Now consider the case where $\tau_i^{t_i}$ is a higher-priority task. Thus, the link is $\tau_w \xrightarrow{-1} \tau_i$. Let again $x(t_i)$ be the value read by $\tau_i^{t_i}$. We must show that $x(t_i) = y(t')$.

The case where the writer is released not more than once before time $t_i$ is identical to the corresponding case in Section 7.7.2. We thus omit it and turn directly to the case where the writer is released twice before $t_i$. Let $c^{t'}$ be the value of `current` that is chosen at time $t'$. Since `current` is not modified between $t'$ and $t$, we have:

$$p^t = c^{t'}.$$

The value $y(t')$ is written in buffer position $c^{t'}$ and this is not modified until $t$, when the writer is released next. At this point, $p^t \neq j^t$, or $c^{t'} \neq j^t$, thus, this position is not over-written by the instance $\tau_w^t$.

`previous` is not modified between $t$ and $t_i$, thus, we have:

$$P^{t_i}[i] = p^t = c^{t'}.$$

`P[i]` is not modified between $t_i$ and $t'_i$, thus, at time $t'_i$, $\tau_i^{t_i}$ reads the value

$$B^{t'_i}[P^{t'_i}[i]] = B^{t'_i}[P^{t_i}[i]] = B^{t'_i}[p^t] = B^{t'_i}[c^{t'}].$$

To show that $\tau_i^{t_i}$ reads the correct value $y(t)$, we must show that the position $c^{t'}$ is not over-written by any instance of the writer until time $t'_i$. Notice that no instance of the writer is released between $t$ and $t_i$, by definition of $t$ and $t_i$. Also, if an instance of the writer is released between $t_i$ and $t'_i$, this instance cannot execute before $\tau_i^{t_i}$ finishes, because it has lower priority than $\tau_i^{t_i}$.

## 7.8 Buffer requirements: lower bounds and optimality of DBP

In this section we study the buffer requirements of semantics-preserving implementations. First, we provide lower bounds on the number of buffers required in the worst case, that is, the maximum number of buffers required for any possible arrival/execution pattern. These lower bounds are equal to the number of buffers used by DBP, thus, the corresponding numbers of buffers are both necessary and sufficient. Second, we show that DBP is using buffers optimally not just in the worst case (i.e., worst arrival pattern) but in any arrival pattern.

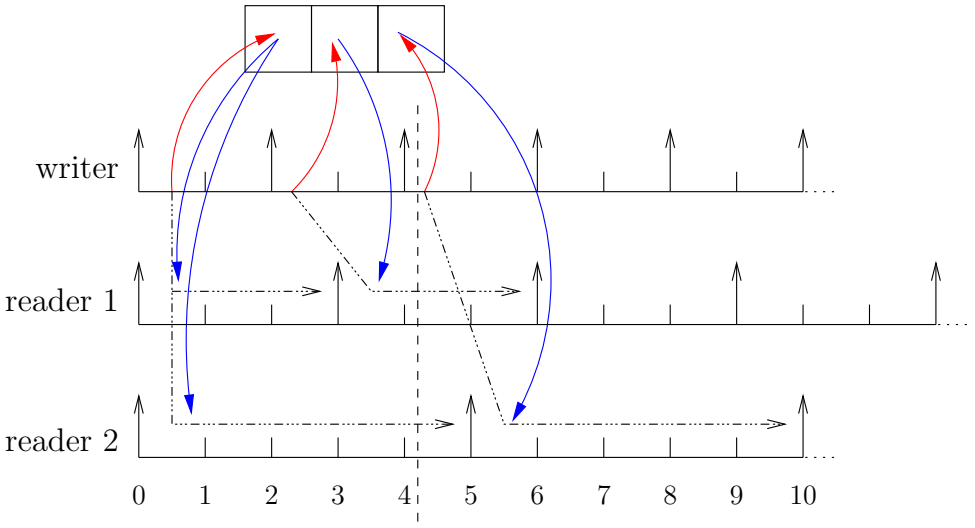



**Figure 7.31:** $N + 1$ buffers are necessary in the worst case ($N$ is the number of readers).

### 7.8.1 Lower bounds on buffer requirements and optimality of DBP in the worst case

We begin this section with a concrete example, for the sake of understanding. Consider the scenario of Figure 7.31, where there is one writer task and two lower-priority reader tasks without unit-delay. The writer $\tau_w$ has period $T_w = 2$ and the readers $\tau_1$ and $\tau_2$ have periods $T_1 = 3$ and $T_2 = 5$, respectively. We assume static-priority, rate-monotonic scheduling [LL73], where priorities are ordered according to the inverse of the periods. That is: $p_w > p_1 > p_2$. In this example, we need $3 = 2 + 1$ buffers. The three buffers are used to store the outputs of the first, second and third occurrences of $\tau_w$, respectively. The first output is needed by the first occurrence of both $\tau_1$ and $\tau_2$. The second output is needed by the second occurrence of $\tau_1$. The third buffer is necessary because when $\tau_w$ starts writing at time 4, $\tau_2$ may not have finished reading from the first buffer yet. Note that in the above example the rate-monotonic assumption is not required, only that the writer has the higher priority is sufficient to generate the described results.

We now provide generalized scenarios and bounds. We consider again the situation of Section 7.5: one writer, $N_1$ lower-priority readers without unit-delay, $N_2$ lower-priority readers with unit-delay, and $M$ higher-priority readers (with unit-delay). Again we let $N = N_1 + N_2$.

First, consider the case $M = N_2 = 0$ (i.e., there is no unit-delay). We claim that $N + 1 = N_1 + 1$ buffers are required in the worst case. Consider the scenario shown in Figure 7.32. There are $N+1$ arrivals of the writer and one arrival of each reader. We assume that when the $(N+1)$-th arrival of the writer occurs, none of the readers has finished execution. This is possible, because the readers have lower priority from the writer and they can be preempted on every new release of it. Moreover, the schedulability assumption is not violated. In the figure we show the *lifetime* of each buffer: for `i`$= 1, ..., N$, buffer `B[i]` is used from the moment of the `i`-th arrival of the writer until the $(N + 1)$-th arrival. A buffer is needed at the last arrival so that the writer does not

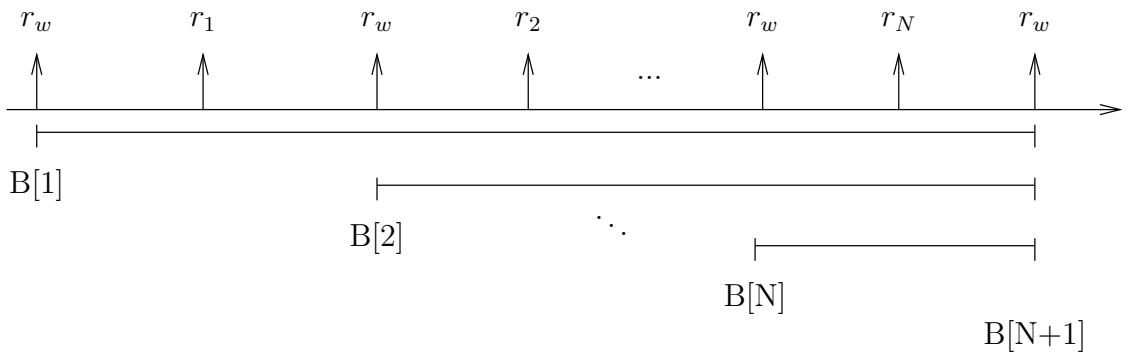



**Figure 7.32:** Worst-case scenario for $N+1$ buffers: $N$ lower-priority readers without unit-delay.

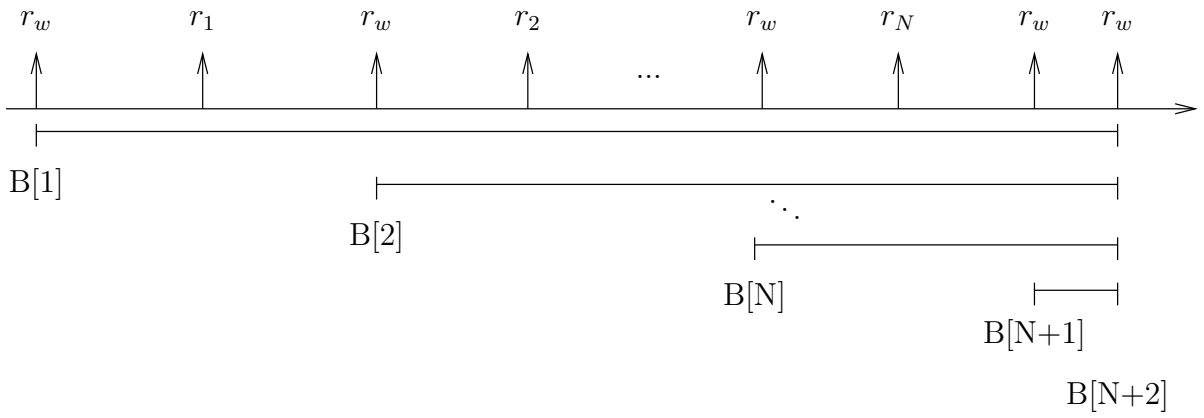



**Figure 7.33:** First worst-case scenario for $N+2$ buffers: $N$ lower-priority readers without unit-delay and at least one higher-priority reader (with unit-delay).

corrupt the data stored in one of the other buffers.

Next, consider the case $M > 0$ and $N_2 = 0$ (i.e., there is a unit-delay). Then, $N_1 + 2$ buffers are required in the worst case. This can be shown using a slight modification of the previous scenario, by adding one more occurrence of the writer at the end: this is shown in Figure 7.33. The last buffer B[$N+2$] is needed because none of the first $N+1$ buffers can be used: buffers B[$1..N$] are used by the $N$ lower-priority readers and buffer B[$N+1$] stores the previous value which may be needed when a higher-priority reader with unit-delay arrives (the latter is not shown in the figure).

Finally, consider the case $M = 0$ and $N_2 > 0$ (i.e., there is again a unit-delay). Then, $N+2$ buffers are required in the worst case, where $N = N_1 + N_2$. A worst-case scenario is shown in Figure 7.34. In the first part of this scenario $N_1$ lower-priority readers without unit-delay arrive, interlaced with $N_1$ occurrences of the writer. This requires $N_1$ buffers. Next, $N_2$ lower-priority readers with unit-delay arrive, interlaced with $N_2 + 1$ occurrences of the writer as shown in the figure. This requires $N_2 + 1$ buffers since the previous values are used by the readers: reader $r'_1$

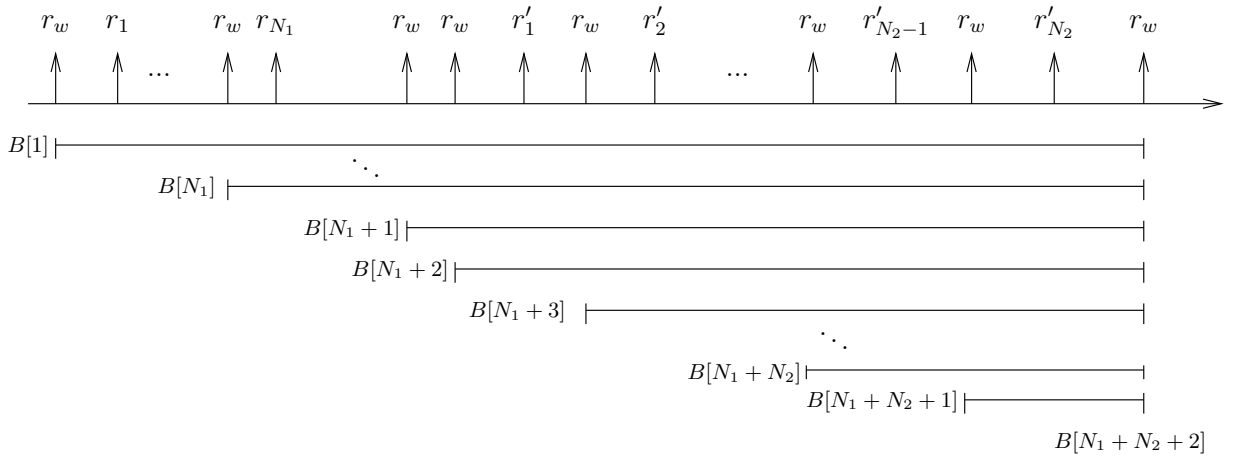



**Figure 7.34:** Second worst-case scenario for $N+2$ buffers: $N = N_1 + N_2$, $N_1$ lower-priority readers without unit-delay and $N_2$ lower-priority readers with unit-delay.

uses $B[N_1+1]$, ..., and reader $r'_{N_2}$ uses $B[N_1+N_2]$. The last writer cannot over-write any of the first $N_1+N_2$ buffers since they are used by readers that have not yet finished. The last writer cannot over-write buffer $B[N_1+N_2+1]$ either, since this stores the previous value which may be needed when a lower-priority reader with unit-delay arrives (the latter is not shown in the figure).

These lower bounds show that DBP is optimal in the worst case, that is, in the "worst" arrival/execution pattern. In the next subsection we show that DBP actually has a stronger optimality property, in particular, it uses buffers optimally in any arrival/execution pattern.

### 7.8.2 Optimality of DBP for every arrival/execution pattern

The protocol DBP is in fact optimal not only in the worst case, but for every arrival/execution pattern, in the following sense:

> for every task graph, for every arrival/execution pattern of the tasks, and at any time $t$, the values memorized by DBP at time $t$ are precisely those values necessary in order to preserve the semantics.

We proceed into formalizing and proving this result.

Let $\rho$ be an arrival/execution pattern: $\rho$ is a sequence of release, begin and end events in real-time (i.e., we know the times of occurrence of each event). We will assume that all writer tasks occur at least once in $\rho$, at time 0, and output their respective default values. This is simply a convention which simplifies the proofs that follow.

For an arrival/execution pattern $\rho$ and for some time $t$, we define $\texttt{needed}(\rho, t)$ to be the set of all outputs of writer tasks occurring in $\rho$ that are still needed at time $t$. Formally, $\texttt{needed}(\rho, t)$ is defined to be the set of all $y$ such that $y$ is the output of THE writer task $\tau_w$ occurring in $\rho$ at some time $t_w$, and one of the following two conditions holds:

1. There exists a link $\tau_w \rightarrow \tau_i$, task $\tau_i$ is released in $\rho$ after $t_w$ and before the next occurrence of $\tau_w$ (if it exists), and $\tau_i$ finishes after $t$.

2. There exists a link $\tau_w \stackrel{-1}{\rightarrow} \tau_i$, there is a second occurrence of $\tau_w$ in $\rho$ at time $t'_w$, where $t_w < t'_w$, $\tau_i$ is released after $t'_w$ and before the next occurrence of $\tau_w$ (if it exists), and $\tau_i$ finishes after $t$.

We assume that outputs $y$ are indexed by the writer identifier and occurrence number, so that no two outputs are equal and $\texttt{needed}(\rho, t)$ contains all values that have been written. This is not a restricting assumption since in the general case the domain of output values will be infinite, thus, there is always a scenario where all outputs are different.

$\texttt{needed}(\rho, t)$ captures precisely the minimal set of values that must be memorized by any protocol so that semantics are preserved. Another way of looking at the definitions above is that $\texttt{needed}(\rho, t)$ contains all outputs whose *lifetime* extends from some point before $t$ to some point after $t$. Notice that $\texttt{needed}(\rho, t)$ is *clairvoyant* in the sense that it can "see" in the future, after time $t$. For instance, $\texttt{needed}(\rho, t)$ "knows" whether a reader $\tau_i$ will occur after time $t$ or not, and if so, whether this will be before the next occurrence of $\tau_w$.

Obviously, a real implementation cannot be clairvoyant, unless it has some knowledge of the arrival/execution pattern. This motivates us to define another set of outputs, denoted $\texttt{maybeneeded}(\rho, t)$. The latter contains all outputs that *may be needed*, given the knowledge up to time $t$. Formally, $\texttt{maybeneeded}(\rho, t)$ is defined to be the set of all $y$ such that $y$ is the output of some writer task $\tau_w$ occurring in $\rho$ at some time $t_w$, and one of the following two conditions holds:

1. There exists a link $\tau_w \rightarrow \tau_i$, such that, if there is a second occurrence of $\tau_w$ in $\rho$ at time $t'_w$, with $t_w < t'_w < t$, then there is an occurrence of $\tau_i$ at time $t_i$, with $t_w < t_i < t'_w$, and $\tau_i$ finishes after $t$.

2. There exists a link $\tau_w \stackrel{-1}{\rightarrow} \tau_i$, such that, if there is a second and a third occurrence of $\tau_w$ in $\rho$ at times $t'_w$ and $t''_w$, with $t_w < t'_w < t''_w < t$, then there is an occurrence of $\tau_i$ at time $t_i$, with $t'_w < t_i < t''_w$, and $\tau_i$ finishes after $t$.

The intuition is that $y$ may be needed because the reader task $\tau_i$ may perform a read operation, say, right after time $t$. It should be clear that for any $\rho$ and $t$, $\texttt{needed}(\rho, t) \subseteq \texttt{maybeneeded}(\rho, t)$.

We want to compare the values stored by DBP to the above sets. To this end, we define $\texttt{DBPused}(\rho, t)$ as the set of all values stored in some buffer `B[i]` of DBP at time $t$, when DBP is executed on the arrival/execution pattern $\rho$, such that `free(i)` is false[5] (recall that the predicate `free` is defined in Figure 7.17).

We then have the following result.

[5] When implementing DBP, there is the option of *pre-allocating* the worst-case number of buffers or allocating buffers *on-the-fly*, that is, during execution, as necessary. This is a usual time vs. space trade-off. To avoid such implementation considerations, we have included in the above definition of $\texttt{DBPused}(\rho, t)$ the requirement that `free(i)` be false, which means that, even if `B[i]` has been pre-allocated, its contents are not needed anymore.

**Theorem 1** *For any arrival/execution pattern $\rho$ and any time $t$,*

$$\texttt{DBPused}(\rho, t) \subseteq \texttt{maybeneeded}(\rho, t).$$

**Proof:**

Consider some $y$ in $\texttt{DBPused}(\rho, t)$. There must be some position `j` such that `free(j)` is false and the value of `B[j]` at time $t$ is $y$. This value was written by the writer $\tau_w$ at time $t_w < t$. We reason by cases:

1. `free(j)` is false because `previous=j`. This means that there is a reader task $\tau_i$ communicating with $\tau_w$ with a unit-delay link $\tau_w \stackrel{-1}{\rightarrow} \tau_i$. We must show that Condition 2 in the definition of $\texttt{maybeneeded}(\rho, t)$ holds. We consider the following cases, depending on how many times $\tau_w$ was released before $t$:

   - $\tau_w$ is not released before $t$. This means that `previous` $=$ `j` $= 1$ and `B[j]` holds the default value $y_0$.
   - $\tau_w$ is released only once before $t$. When this happens, `previous` is set to `current` which is equal to 1, since this is the first release of $\tau_w$. Thus, again `B[j]` holds the default value $y_0$.
   - $\tau_w$ is released at least twice before $t$, and the last two times where at $t_w < t'_w < t$. At $t'_w$, `previous` is set to `current` which equals `j` at that point. Thus, `B[j]` holds the value $y$ written by the instance of $\tau_w$ released at $t_w$.

   In none of the three cases above there is more than one occurrence of $\tau_w$ after $t_w$, thus Condition 2 in the definition of $\texttt{maybeneeded}(\rho, t)$ holds.

2. `free(j)` is false because there is some $i \in [1..N_1]$ such that `R[i]=j`. This means that the reader $\tau_i$ communicates with $\tau_w$ via a link without unit-delay, $\tau_w \rightarrow \tau_i$. Since `R[i]`$\neq$`null`, $\tau_i$ is released at least once before $t$ and it has not finished at time $t$. Suppose $\tau_i$ is released last at time $t_i < t$. When this happens, `R[i]` is set to `current` which equals `j` at that point. Condition 1 in the definition of $\texttt{maybeneeded}(\rho, t)$ holds since $t_w$ is the last occurrence (if any) of the writer before time $t_i$ and $\tau_i$ finishes after time $t$.

3. `free(j)` is false because there is some `i`$\in [N_1 + 1..N_1 + N_2]$ such that `R[i]=j`. This means that the reader $\tau_i$ communicates with $\tau_w$ via a link with unit-delay, $\tau_w \stackrel{-1}{\rightarrow} \tau_i$. This case is similar to Case 1 above.

■

The above result shows that DBP never stores redundant data, only data that may be needed. In the absence of any knowledge about the future (which is unknown if arrival/execution patterns are not known), this is the best that can be achieved, if we are to preserve semantics. However, we can also show that DBP is not too far from the ideal case.

### 7.8.3 Exploiting knowledge about future task arrivals

As we have seen in Section 7.5.3, the *a priori* knowledge of the release times of the tasks can be used to build a static schedule where DBP can refer to for buffer assignments during execution. This static schedule, however, is not sufficient to make DBP optimal in the number of buffers used. This is because DBP is not clairvoyant, as explained before, therefore, it does not exploit knowledge about future task arrivals, even when such knowledge is available. In this section we describe how such knowledge can be used. Let us first explain the technique in the multi-periodic case.

We equip DBP with a predicate *isNeeded(t)* which is true when the value produced by the writer $\tau_w$ at time $t$ is needed by a forthcoming reader. This predicate can be computed based on the periods of writer and reader tasks. For this, we also need the function $l(i,t)$, defined below:

$$l(i,t) = \lfloor \frac{\lfloor t/T_i \rfloor \, T_i}{T_w} \rfloor \, T_w \tag{7.2}$$

where $T_w$ is the period of the writer task $\tau_w$ and $T_i$ is the period of the reader task $\tau_i$. $l(i,t)$ is equal to the last time that $\tau_w$ was released before the last arrival of $\tau_i$ before $t$. Intuitively, this function shows which data of the writer is needed by reader $i$ if this reader is to be executed at time $t$.

Then we can define the predicate *isNeeded(t)* as follows:

$$isNeeded(t) = \exists t' > t. \exists i. [(\exists \tau_w \rightarrow \tau_i \wedge t = l(i,t')) \vee (\exists \tau_w \stackrel{-1}{\rightarrow} \tau_i \wedge t = l(i,t') - T_w)] \tag{7.3}$$

We can now modify DBP as shown in Figure 7.35, in order to avoid assigning buffers to cases where the output of the writer is not needed.

*Writer*:

- When $\tau_w$ is released on time $t$:
  `if isNeeded(t) then`
  `previous := current`
  `current :=` some `j`$\in$`[1..`$N$`+2]` such that `free(j)`
  `else`
  `previous := current`
  `current := null`
  `endif`
- While $\tau_w$ executes and `current`$\neq$`null`, it writes to `B[current]`.

**Figure 7.35:** The protocol DBP.

A similar technique can be applied to other cases apart from multi-periodic where the order of task arrivals is known. The difference will be that the predicate `isNeeded(t)` must be

defined in a different way: when a writer is released at time $t_w$, we check if there is a release of a reader between time $t_w$ and time $t'_w - 1$, where $t'_w$ is the next release of writer after $t_w$. If there is such a release of a reader, then `isNeeded(`$t_w$`)=true` and the standard procedure is followed.

**is the above redundant???**

In addition to the previous, as we saw in Section 7.5.3, DBP protocol can be applied to tasks with known arrival patterns as well. In that case the predicate `isNeeded(t)` is given by a known function with respect to relative arrivals of readers and the writer; when a writer is released at time $t_w$, we check if there is the release of a reader between time $t_w$ and time $t'_w - 1$, where $t'_w$ is the next release of writer after $t_w$. If there is such a release of a reader, then `isNeeded(t)=true` and the standard procedure is followed.

## 7.9 Related Work

Before we conclude this Chapter, we discuss a number of related works. First we present the code generation method that the tool REAL-TIME WORKSHOP applies for SIMULINK and its current limitations. Next, we discuss the paper [BFMSV05] extensively, since it includes the closest work to ours. Finally, we discuss a number of other related works.

**Code generation for SIMULINK/STATEFLOW models using REAL-TIME WORKSHOP.** The documentation of *MathWorks*' code generator REAL-TIME WORKSHOP claims that the tool builds code that reproduces the deterministic behavior of the model, provided tasks are periodic and periods are multiples of each other[6]. The documentation does not describe how this is achieved, however, evidence can be found in some restrictions imposed on *multi-rate* SIMULINK diagrams. For instance, SIMULINK requires that a *rate-transition* block must be inserted every time we have different rates between two connected blocks.

A Rate-transition block behaves as a *unit-delay* block, in the case where a "slow" writer communicates data to a "fast" reader. In this case the block executes in the rate of the writer but with the priority of the reader. The block will copy the output of the writer and provide it to the reader upon his execution. Also a delay of 1 will be added in that communication, i.e., the reader will always read the previous last result of the writer. On the other hand if a "fast" writer sends data to a "slow" reader the rate-transition block acts like a *zero-order hold* block , i.e., it is executed in the rate of the reader but with the priority of the writer, thus having always the correct value needed by the writer.

However, this solution imposes the use of multi-periodic tasks and moreover that the reader and the writer always have multiple periods, otherwise this solution fails. Note also, that the above solution implies that the scheduling is done according to the *rate monotonic* priority scheme [LL73], where the faster one task is executed, the higher priority it has.

[6] Quoting from Section "Mapping Model Execution to the Target Environment" of [Mat]: "A correctly executing application will generate deterministic results that are identical to the results produced by the model in simulation. To achieve correct execution, the model's sample rates must be mapped into corresponding tasks executing in the target environment." See the section titled "Models with Multiple Sample Rates" of the Real-Time Workshop user guide, available at http://www.mathworks.com/access/helpdesk/help/toolbox/rtw/ug/.

Our solution is more general since it applies also to event-triggered applications, to arbitrary priority assignments, as well as to the EDF scheduling algorithm. Moreover, our solution is memory-optimal when extended to many readers and writers.

**Memory bounds on the implementation of synchronous programs.** [BFMSV05] presents work similar in spirit to ours, however, with a number of differences. The main focus of [BFMSV05] is to provide upper bounds on the memory required by any inter-task communication protocol so that the synchronous semantics are preserved. Our emphasis is instead on the development of communication schemes that achieve optimal memory bounds.

More specifically, the differences between the two works are the following. First, regarding the setting:

- We work with a known scheduling policy (static-priority or EDF). In [BFMSV05], no assumption is made about the scheduling policy.
- We require no knowledge of the minimum inter-arrival time (MIT) or deadline of tasks. However, we do require schedulability. In [BFMSV05], it is assumed that the MIT and deadline of each task is known. Schedulability is also assumed, in the sense that the deadline of each task is respected. The deadline of a task can be greater than its MIT, which means that the schedulability assumption in [BFMSV05] is weaker than ours.
- We assume that all tasks run on a single processor. In [BFMSV05], no a-priori assumption is made about the execution platform.

Then, regarding the memory requirements:

- We provide protocols that use buffers optimally.
- The bounds provided in [BFMSV05] are generally not tight. For example, in a *static-priority* (SP) setting with one writer and $N = 2$ readers, where both readers have lower priority than the writer, where the periods are $T_w = 2, T_1 = 3, T_2 = 5$, and the deadlines are equal to the periods, we require 3 buffers whereas the upper bound calculated by the formulas provided in [BFMSV05] is 4.

Finally, regarding the communication schemes:

- DBP is *lock-free*, meaning that tasks do not block on reads or writes: the only thing that can suspend execution of a task is preemption by another, higher-priority task. DBP requires *atomic* manipulations of global pointers upon task releases. These can be handled by the operating system or by some special interrupt-handling routine with the highest priority.
- [BFMSV05] considers single-processor, "multi-processor" (many processors with centralized pointer manipulations) or "distributed" (many processors with decentralized pointer manipulations) implementations. For single-processor implementations, both lock-free and locking methods are considered. In the lock-free methods, the assumption is that

scheduling ensures that the producer has always higher priority than the consumer. This implies that there can be no cycles in the graph of producers/consumers and that this graph is topologically ordered in order to assign the priorities. The main impact of this assumption is that a consumer task which comes after a long chain of producers will have a very low priority, thus, it cannot handle "urgent" events. In contrast, in our work we make no assumption on the relative priorities of producer and consumer, provided that a unit delay is present when the producer has lower priority than the consumer.

In summary, one can say that the setting of [BFMSV05] is more general than ours, however, it also requires more knowledge (minimum inter-arrival times, deadlines). Our protocol has optimal memory requirements while the upper bounds provided in [BFMSV05] are not tight (this is to be expected given that their setting is more general).

**Other Related Work.** [LM87] introduces the *synchronous dataflow* model (SDF) and provides methods for static scheduling and code generation on single-processor or multi-processor architectures. This work has been extended in a number of directions, including buffer optimizations (e.g., see [BML96, MB04]). SDF can be viewed as a subclass of the model we consider in this paper, in the sense that only multi-periodic designs can be described in SDF. On the other hand, SDF descriptions are more high-level and must generally be *unfolded* into a more basic model such as ours. A major difference with our work, however, is that [LM87, BML96, MB04] aim for static, cyclic schedules, whereas we aim for multi-task applications that use dynamic, preemptive scheduling.

[Nat06] considers the problem of minimizing the cost of adding unit-delay or zero-order-hold blocks in SIMULINK diagrams. The main difference with our work is that the focus of [Nat06] is cost minimization and not preservation of semantics. Indeed, by adding or removing blocks as the above, the semantics generally change. In contrast, we start from a fixed set of such blocks and do not attempt a modification. We only perform optimizations at the implementation level and not at the design level.

[RB04] proposes a method of distribution of synchronous programs and preserves the ideal semantics by introducing delays in the early (high-level) design stages.

Our work is also related to a set of papers that propose lock-free inter-task communication schemes, for instance [CB97, HPS02]. Although our methods are lock-free (only manipulations of pointers are atomic, writes and reads need not be), it is different from the protocols proposed in the above works. The latter preserve the integrity and often also the "freshness" of data, meaning that the reader consumes the latest complete value produced by the writer. This value does not always correspond to the value defined by the zero-time semantics. Another difference, which is the basic characteristic of our work, is that the protocols are based on pointer manipulations that happen upon task release, and not task execution, as is the case of the above protocols.

We should also note that our work has different objectives from the research done in the context of real-time scheduling theory (e.g., see [LL73, HKO+93, ABD+95, SSRB98]). Real-time scheduling theory is concerned with checking schedulability of a set of tasks in various settings. Our concern is not schedulability, but preservation of semantics. We *assume* that the system is

schedulable (something that can be checked using existing scheduling techniques such as those in the works cited above) and we develop preservation schemes that rely on this assumption.

## 7.10 Conclusions

In this Chapter we studied the problem of semantics-preserving implementations of synchronous programs on single-processor, multi-tasking execution platforms with preemptive scheduling policies such as static-priority or EDF. We proposed a set of buffering protocols for inter-task communication that preserve the synchronous semantics, that is, guarantee that the behavior of the executed code will be the same (in terms of streams of output values) to the behavior of the synchronous program. We proved this using model-checking as well as "manual" techniques. We also proved that our generalized protocol (for the many writers-many readers case), DBP, is optimal with respect to memory requirements.

# Chapter 8

# Conclusions and Perspectives

The success of the SIMULINK/STATEFLOW toolkit is largely a result of the fact that a designer can model the system coupled with the environment and simulate its execution. This success shows that the industry of embedded control systems needs a complete approach starting from the high-level conceptualization of the system until the final implementation in the target platform.

Such an approach raises a number of important issues, especially in the domain of safety-critical applications (avionics and automotive being two examples). We need model-checking capabilities for verification purposes and when moving towards the execution platform, we need an implementation methodology that preserves properties of the high-level design. Moreover, efficiency of implementations in terms of memory usage or time performance is a must.

This is exactly the problem that this thesis treats and proposes solutions to. Indeed we propose a complete chain from top to bottom for the design and implementation of embedded control systems. This is a three-layered approach, starting from the design using a high-level modeling and simulation tool, in particular SIMULINK/STATEFLOW, and going down to the execution in a single-processor, multi-tasking preemptive platform, while preserving the synchronous semantics of the model all-along. We use LUSTRE as an intermediate language, because it provides a formal modeling framework, rich with tools such as model-checkers, test-generators and code-generators.

This work opens many directions for future research. Some general directions, in the context of the overall model-based design effort at VERIMAG, are the following:

- Enlarging the set of high-level models in the picture of Figure 2.1. We have considered SIMULINK and STATEFLOW. Another high-level modeling formalism (or rather, collection of formalisms) is the Unified Modeling Language (UML) and its various extensions, such as SysML[1]. This language, coming from the software engineering domain rather than control, addresses somewhat different concerns, which explains the interest of the industry in it. At the same time, UML brings different challenges to an automated, semantics-preserving implementation process, and this, independently of whether LUSTRE is considered or not as an intermediate point.

[1] see http://www.omgsysml.org/

- Enlarging the set of execution platforms. We now understand relatively well the implementation of synchronous models on single-processor platforms, single-tasking or multi-tasking. Some initial work has been also done for distributed platforms, such as the Time-Triggered Architecture or TTA [CCM+03], which is a synchronous execution platform, and for multi-periodic applications. A lot remains however to be done in this direction. For instance, we still need to cover loosely synchronous [BCG+02] or asynchronous architectures with preemptive scheduling for more general task arrival patterns. Also we can study the implementation in a distributed architecture equipped with a CAN or Flexray bus, or a network of different buses connected through gateways.

Some directions more specifically related to the work done in this thesis are the following:

- Our translation method from SIMULINK to LUSTRE is restricted to the discrete-time part of SIMULINK. A natural extension would be to consider continuous-time SIMULINK as well. This is not a trivial problem, as it includes in essence the entire issue of relating continuous-time and discrete-time systems, with the theory of sampling and other major control-theoretic results. Some recent work in this direction has been done in [CB02, KC06].

- Both translations of SIMULINK and STATEFLOW are "one-way", in two senses: first, we do not currently translate LUSTRE into SIMULINK (or STATEFLOW), and second, we do not map the results provided by LUSTRE-based tools such as the LESAR model-checker back to the original model. Both limitations are serious, in particular the second one, since it obliges the user to "look under the hood", so to speak, and inspect the results of the analysis at the LUSTRE level. This requires familiarity with the LUSTRE language, which SIMULINK/STATEFLOW users often lack.

- Our study of static analysis methods for STATEFLOW is only preliminary. In particular, the techniques proposed in Chapter 5 have not been implemented. They should be, as well as tested on a number of case studies. This would provide feedback in order to validate and improve these methods.

- Another, more ambitious goal, is to develop dynamic model-checking algorithms for Stateflow, that are specific for the semantical problems of STATEFLOW (e.g., stack-overflow or confluence). Model-checking for STATEFLOW is generally undecidable and so is checking these properties, however, optimizations with respect to general model-checking algorithms are probably possible.

- The prototype tool S2L that implements the algorithms studied in Chapter 4 can be further extended to translate more blocks from the SIMULINK library that initially were not considered. In addition to that, newer releases of SIMULINK add new features and change the input file representation of the model. Further study should be done in such additions and incorporate those changes in the tool.

- The buffering protocols we proposed in Chapter 7 rely on support from the operating system, in particular regarding the management of pointers to the buffers upon arrivals of task-triggering events. Methods to implement this mechanism on current RTOSs in an efficient way (i.e., with minimal overhead) should be studied. One possibility for implementing the mechanism without changing the RTOS kernel is to build special high-priority interrupt handling routines that are executed upon arrival of the triggering events and manipulate the pointers. This solution should be evaluated with respect to others, in particular those that rely on a modification of the kernel.

- In Chapter 7, we proposed only a simple and limited method to decompose synchronous programs into tasks. This problem should be studied thoroughly.

- In the context of multi-tasking implementation of synchronous programs on preemptive execution platforms, we can consider relaxing the schedulability assumption, to allow multiple instances of the same task to be active at the same time.

- More generally, our multi-tasking implementation scheme relies on external methods and tools for scheduling and schedulability analysis. We can consider connecting the two in a "feedback loop", where information from the scheduler and schedulability analyzer is provided to the code generator, in order to optimize the final solution.

# Appendixes

# Appendix A

# Help messages by the S2L and SF2LUS tools

The help message provided by the S2L tool when this one is invoked from command line is the one appearing in the Figure A.1

While the output of the SF2LUS program gives the following help message:

```
Simulink to Lustre Jan05
Usage: s2l <OPTIONS>... <FILE.mdl>
Translates a simulink model to Lustre

where options can be
-p,--pollux             Generates Lustre code for the Pollux
                        Lustre compiler
-d,--debug              Provides debug information to standard
                        output
-m                      Generates a MainNode that folds the
                        entire model
-mp, --monoperiodic     All the flows in the lustre program will
                        have the same clock
--side-effects          In case S-Function produce side effects
                        they are distinguished external function
                        calls
--choose-system         Choose the subsystem to tranlslate if not
                        the entire model.
-o <fileName>           The lustre output file name
--sf-lustre-file        The Lustre file of the SF
-f,--fullnames          The names of nodes and certain variables
                        correspond to their path
--varnames              Output the correspondence of the variable
                        names in the lustre file
-xml                    Translates the model to XML and exits.
-v,--version            Prints the version of the program
--help                  Display this help and exit

report bugs at          Christos.Sofronis@imag.fr
```

**Figure A.1:** The help message provided by the S2L tool.

```
Stateflow to Lustre (c) VERIMAG 2004

  Convert Stateflow into Lustre.

Syntax:

  sf2lus <options> file.mdl

Bug reports and enquiries to: "Paul Caspi" <Paul.Caspi@imag.fr>

Options:
  -r13                    Matlab version 13
  -r14                    Matlab version 14 (default)
  -kw <str>               Add a keyword to the keyword identifier
                          list
  -nkw <str>              Remove a keyword from the keyword
                          identifier list
  -paths                  Use full path names for states
  -no_paths               Do not set -paths automatically
  -I <dir>                Append a directory to the search path
  -include <file>         Add a file to be included
  -o <file>               Name of output file, (default: stdout)
  -mws <file>             Name of Matlab workspace emulation file
  -margin <int>           Set the margin for formatted output
  -max_indent <int>       Set the maximum indent for formatted
                          output
  -text_limit <int>       Limit output strings to this number of
                          characters
  -pollux                 Use Pollux modifications (default)
  -nbac                   Use Nbac modifications
  -reluc                  Use Reluc modifications
  -scade                  Use Scade modifications
  -names                  Use state names in variables
  -ids                    Use state ids in variables
  -names_ids              Use both names and state ids in variables
  -long_names             Use unabrreviated names
                          (eg. "s" -> "state")
  -no_self_init           Top level graph does not provide
                          initialization
  -ess <int>              Event stack size
  -sends                  Enable sends to specific state
                          (events become ints)
```

```
-errstate                 Add error output variable
-junc_states              Treat junctions as states
-create_missing           Add missing data to data dictionary
-missing_scope <scope>    Scope for missing data
                          (default: INPUT_DATA)
-missing_datatype <type>  Data type for missing data
                          (default: double)
-no_constants             Omit workspace constants from output
-emulate_time             Provide internal time value
-start_time <float>       Start time for emulated time
-time_increment <float>   Time increment for emulated time
-real_time                Provide real time value (in external
                          C code)
-varprefix <str>          Prefix all variables (for namespace
                          conflict avoidance)
-prefix <str>             Prefix all names (used for comparisons
                          with lesar)
-suffix <str>             Suffix all names (used for comparisons
                          with lesar)
-observe <expr>           Add observer node for given expression
-no_observers             Don't read observer file
-consistency              Add state consistency observer
                          (sets -states_visible)
-counters                 Add loop counters to junctions
                          (sets -junc_states)
-unroll                   Unroll loops according to loop counters
-trace                    Add trace output
-trace_inputs <int>       Number of inputs to add to trace output
-trace_locals <int>       Number of locals to add to trace output
-export_cvs               Export condition variables (set if
                          function call events)
-states_visible           Make state variables visible for toplevel
                          graph
-temps_visible            Make temporary variables visible for
                          toplevel graph
-stubs_visible            Make stub nodes visible in output
                          (won't compile)
-locals_visible           Make chart locals outputs
-no_typecheck             Do not typecheck generated nodes
-no_sequence              Do not sequence generated nodes
-no_normalize             Do not normalize generated nodes
```

```
-input_bools_ints       Transform input booleans into ints
                        (for luciole)
-write_now              Write output as generated (debug)
-g                      Enable debug printouts
-gp                     Enable parser debug printouts
-v                      Set debug level
-help                   Display this list of options
--help                  Display this list of options
```

**Figure A.2:** The help message provided by the SF2LUS tool.

# Appendix B

# One-to-one translation of a SIMULINK model to XML

As mentioned in Chapter 4, the first step of the S2L tool is the translation of the SIMULINK model to an equivalent (in the sense of the information contained in both representations) XLM document. This is done with the use of the `fr.verimag.mdl2xml` package as we saw in the tool's architecture in Figure 6.1. In this Appendix we demonstrate an example for this translation.

Figure B.1 shows a SIMULINK model file. However, since the length of that file (and therefore of the translated XML file) is too big we provide only some segments of them. Figure B.2 shows XML file that represents the SIMULINK model file (and that will be fed to the S2L tool to continue with the translation to LUSTRE).

We can see clearly from this example that the produced XML file does not contain information that are irrelevant to the translation to LUSTRE, such the `ScreenColor` and `PaperOrientation` properties of the source file, or the `Position` of the blocks. However, this is an option that can be disabled and produce an XML that is completely identical in terms of information.

```
Model {
  Name                    "CombinatorialLogic"
  Version                 5.0
...
  System {
    Name                    "CombinatorialLogic"
    Location                [182, 292, 787, 577]
    Open                    on
    ModelBrowserVisibility  off
    ModelBrowserWidth       200
    ScreenColor             "white"
    PaperOrientation        "landscape"
    PaperPositionMode       "auto"
    PaperType               "usletter"
    PaperUnits              "inches"
    ZoomFactor              "100"
    ReportName              "simulink-default.rpt"
    Block {
      BlockType               Inport
      Name                    "Set"
      Position                [35, 88, 65, 102]
      SampleTime              "1"
      Interpolate             off
    }
...
    Line {
      Labels                  [0, 0]
      SrcBlock                "Reset"
      SrcPort                 1
      DstBlock                "Mux"
      DstPort                 2
    }
...
  }
}
```

**Figure B.1:** The initial model file of the example of the translation to XML.

```
<?xml version='1.0' encoding='UTF-8'?>
<xml>
<Model>
    <property name="Name" value="CombinatorialLogic"/>
    <property name="Version" value="5.0"/>
...
    <System>
        <property name="Name" value="CombinatorialLogic"/>
        <Block>
            <property name="BlockType" value="Inport"/>
            <property name="Name" value="Set"/>
            <property name="Position" value="[35␣88␣65␣102]"/>
            <property name="SampleTime" value="1"/>
            <property name="Interpolate" value="off"/>
        </Block>
...
        <Line>
            <property name="SrcBlock" value="Reset"/>
            <property name="SrcPort" value="1"/>
            <property name="DstBlock" value="Mux"/>
            <property name="DstPort" value="2"/>
        </Line>
...
    </System>
</Model>
</xml>
```

**Figure B.2:** The resulting XML file.

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