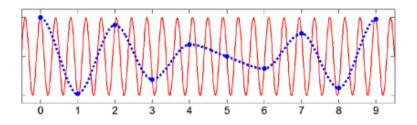
Reasoning About Sound Programs

Emilio Jesús Gallego Arias

Joint work with O. Hermant & P. Jouvelot MINES ParisTech, PSL Research University, France

Rennes, 15 Avril 2015



Some Music DSLs

* DARMS

DCMP

- # #CED
- a Adapie
- s AML
- # AMPLE
- # Avenie
- # Autoklane
- · Bang
- · Carron
- · CHANT
- # Church
- · CLCE
- * CMIX
- # Consuir
- CMUSIC
- Common Line Munic
- · Common
- · Common Manie
- Notation · Cround
- Caballa

- # DARK
- # Elasty
- a EsAC
 - # Exterpes
 - · Extension
 - a Faunt
 - a Figures

 - H Finnes
 - # FOR
 - # FORMES

 - # GUIDO
 - · HARP
 - a Hashere

- # 150V

- · FORMULA
- # Fugue
- # Gibber
- · GROOVE

- a HMASL

 - invokator * KERN

 MU58 MUSCMP

LPC

Mars

a Max

MASC

· MidiLing

· Midil.ogo

MODE

* MOM

a Mont

* M5X

MUS10

MuneData

· MUSIC 10

MUSIC 11.

MUSIC 360

MUSIC 48

· MUSIC 4F

· MUSIC

485

MunES

. PLACON

PLAYI

PLAY2

POCO

POD6

POD7

· PROD

= PWGL

a Ravel

SALIERI

· SCORE

ScoreFile

SCRIPT

SIREN

· SMDL

· SSP

ST

555P

· Supercollider

Symbolic

SMOKE

= Puredata

a PMX

MCL

MUSIC

HI/IV/V

· MusicLogo

Music1000

MUSIC7

Municipal

MUSIGOL

MusicXML

a Munisters

NOTELIST

OpenMusic

· Organum1

· Outperform

Overtone

a Patchmerk

· PE

· PILE

1Phu

Nymist

. OPAL

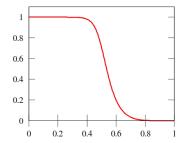
a MIFE

Some Music				0
	* DARMS			
	# DCMP			
# ACED	# DMIX	* LPC		PLACOMP
a Adagio	# Elady	# Mars	# MCL	R PLAYI
# AML	# EsAC	# MASC	# MUSIC	* PLAY2
# AMPLE	# Esterpes	# Max	III/IV/V	# PMX
# Arctic	* Extempore	* MidiLisp	 MusicLogo 	· POCO
# Autoklang	a fam	# MidLogs	 Music1000 	# POD6
		and the second sec	 MUSIC7 	POD7
* Bang	Cofty		ification	ROD
# Canon	Solly	vare ver	Incalior	redata
# CHANT	# FOR		 MUSICAML 	- WGL
# Chuck	* FORMES	* MSX	# Musictes	 Ravel
# CLCE	· FORMULA	* MUS10	# MIFF	# SALIERI
* CMIX	# Fugue	* MU58	# NOTELIST	· SCORE
# Crussic	* Cibber	# MUSCMP	# Nymuist	# ScoreFile
· CMUSIC		# MuseData	. OPAL	# SCRIPT
· Common	* GROOVE	# MusES	# OpenMusic	. SIREN
Lisp Munic	# GUIDO	· MUSIC 10	* Organum1	. SMDL
# Common	* HARP	· MUSIC 11	a Outperform	. SMOKE
Music	# Hashore	# MUSIC 360	# Overtone	. SSP
Common	# HIMSL	# MUSIC 48	* PE	# 555P
Munic	# INV	. MUSIC	a Patchwork	# ST
	a invokator	487	· PILE	supercullider
# Capund	* KERN	. MUSIC 4F		 Symbolic
+ CuberRand			# Pla	a synemic

Some Music * 4CED * Adagio * Adagio * Adagio * Adagio * Antelia * Anteliang * Bang * Canon * CHANT * Churk	DARAS DOMPA DOMP DOMP DOMP DOMP DOMP DOMP DOMP DOMP DOMP DOMPA DOMPA	· LPC · Mass · M	 Muser A MI 	WGL
# CLCE # CMIX	What	t is oui	r gai	n?
 Crivitic CMUSIC Common Lisp Music Common Music Common Music Common Music Coheritori 	 a Gibber a GROOVE a GUBO a HARP a HARP a HARS, a HMS, a HNV a invokator a KERN 	 MoneData MonES MUSIC 10 MUSIC 11 MUSIC 13 MUSIC 48 MUSIC 48 MUSIC 4F 	 Preparent OpenMusic OrganumI Outperform Overtone PE Patchmork PHE PA 	 SCRIPT SIREN SMOL SMOKE SSP SSP ST Supercalider Symbolic

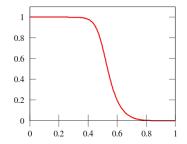
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smooth_n = $(1 - c) \cdot x_n + c \cdot \text{smooth}_{n-1}$



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What would we like to know about it?

smooth_n = $(1 - c) \cdot x_n + c \cdot \text{smooth}_{n-1}$

Natural questions are:

- Frequency response;
- Stability;
- Linearity/Time Invariance.

Standard DSP theory gives answers.

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Paradigm shift!

Faust

- Functional PL for digital signal processing.
- Synchronous paradigm, geared towards audio.
- Programs: circuits/block diagrams with feedback.
- Semantics: streams of samples.
- Efficiency is crucial.
- Created in 2000 by Yann Orlarey et al. at GRAME.
- Mature, compiles to more than 14 platforms.

Faust's Ecosystem

Users:

- Grame: Multiple projects, main developer.
- Stanford: Class/books on signal processing, STK instrument toolkit, Faust2android, Mephisto...
- Ircam: Acoustic libraries, effects libraries,...
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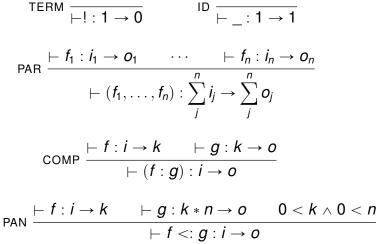
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It has its market! Much easier than dwelling into C. Recent Events:

- Faust day at Stanford, LAC 2015.
- ► Faust program competition (€2,000).
- FEEVER project :)

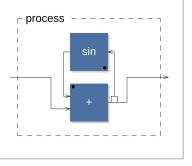
Syntax and Well-Formedness



Feedback

$$\texttt{FEED} \; \frac{\vdash f: o_g + i_f \rightarrow i_g + o_f \qquad \vdash g: i_g \rightarrow o_g}{\vdash f \sim g: i_f \rightarrow i_g + o_f}$$

Diagram for + \sim sin:

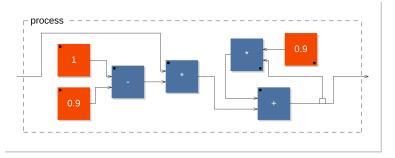


Back to the Filter

smooth_n =
$$(1 - c)x_n + c \cdot \text{smooth}_{n-1}$$

Using Faust:

 $smooth(c) = *(1-c) : + \sim *(c)$



[For c = 0.9]

Feedback Delay Network

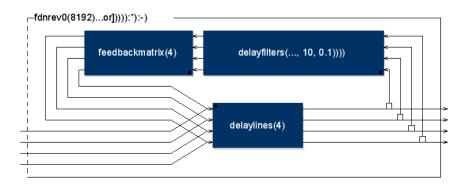
```
fdnrev(N, dp, freqs, durs, loopgainmax)
  = delaylines ~ (delayfilters : feedbackmatrix)
where
  delaylines = rep(N,i,delay(dp[i])));
  delayfilters = rep(N,filter(freqs,durs));
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Spectral processing may open a new gap from all of those! Some related DSL: VOBLA, Ziria, Halide, Darkroom, Julia.

DSP & Faust

- Real-time Linear Processing.
- Real-time Non-linear Processing.
- Frequency Domain Processing.
- Non-necessarily causal.
- Filters, Feedback Networks, Interpolation.
- Windowing!
- Numerical issues.
- Nyquist/precision/aliasing.

- Model checking/automata.
- Program analysis/logics.
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Use mechanized techniques to ensure correct behavior.

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Problems with Audio:

bad sound, stability/glitches, under/overflows, time, safety/security, remote distribution.

We need more!

A Case Study: Stability

Test-bed: use Coq Coq is a theorem prover that provides very strong evidence as compared to Mathlab, etc... A Case Study: Stability

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Smooth is stable when $c \in]0, 1[$. Formally:

 $\forall i \in [a, b], c \in]0, 1[\rightarrow smooth(c) i \in [a, b]$

Let's build a mechanized constructive proof.

1. Define the syntax of Faust inside Coq.

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- 5. Verify!

Mechanized Semantics for Streams

- Coinductive semantics [Boulmé, et al]: problematic.
- Didn't consider PACO, etc....
- ▶ Our wish: Sequences S of a base type **R** [Auger2013]

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Soundness needs stronger semantics (also [Guatto2014]):

$$\llbracket \vdash f: i \to o \rrbracket^n : \llbracket \underbrace{\mathbb{R} \times \ldots \times \mathbb{R}}_{i} \rrbracket^n \to \llbracket \underbrace{\mathbb{R} \times \ldots \times \mathbb{R}}_{o} \rrbracket^n$$

Index by number of steps; equality of streams more intensional wrt to $(\mathbb{N} \to \mathbf{R})$.

The Second Piece: Real Analysis

What about the base type R?

- Reals not in Mathcomp algebraic structures good enough for most of our experiments so far.
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Not practical.

To remedy this, we define a *program logic* for sample-level properties.

Sampled-Level Predicates

Definition (Sample-Level Property)

A property $P : \mathbb{S} \to \mathbb{B}$ is sample-level if there exists a characteristic predicate $\varphi : \mathbf{R} \to \mathbb{B}$ such that for all streams *s*:

$$P(s) \iff \forall n.\varphi(s[n])$$

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Boundedness $x \in [a, b]$ is a sample-level property! Properties can be made sample-level by self-composition, e.g: ratio:

$$f \Rightarrow \langle f, f' \rangle : /$$

We can also prove this way equivalence of filter implementation.

A Sampled Logic

Definition (Sampled Judgment)

Given two characteristic predicates φ, ψ , we write

 $\{\varphi\} \ \mathit{f} \ \{\psi\}$

"for all input i meeting φ , the fi satisfies ψ ."

Example The stability judgment for smooth is written as:

 $\{x \in [a, b]\}$ smooth $\{x \in [a, b]\}$

Rules for The Sampled Logic

$$\frac{\forall i_1, i_2, (\varphi_1(i_1) \land \varphi_1(i_2)) \Longrightarrow \psi(i_1 + i_2)}{\{\varphi_1, \varphi_2\} + \{\psi\}} Prim$$

$$\frac{\{\varphi\} f \{\theta\} \quad \{\theta\} g \{\psi\}}{\{\varphi\} f : g \{\psi\}} Comp$$

$$\frac{\models \psi(x_0) \quad \{\theta, \varphi\} f \{\psi\} \quad \{\psi\} g \{\theta\}}{\{\varphi\} f \sim g \{\psi\}} Feed$$

Soundness of the Logic Definition (Validity)

 $\llbracket \{\varphi\} \ f \ \{\psi\} \rrbracket \equiv \forall i.(\forall t.\varphi(i(t))) \implies (\forall t,\psi(\llbracket f \rrbracket)(i)(t))$

Theorem (Soundness) For any program f of type $i \rightsquigarrow o$, if

$$\{\varphi_1,\ldots,\varphi_i\} f \{\psi_1,\ldots,\psi_o\}$$

is derivable then,

$$\llbracket \{\varphi_1,\ldots,\varphi_i\} f \{\psi_1,\ldots,\psi_o\}\rrbracket$$

is valid.

Stability Proof for Smooth

$$\frac{\square}{\{I_{ab}\} * (1-c) \{I_{ab\overline{c}}\}} \qquad \frac{\square}{\{I_{abc}, I_{ab\overline{c}}\} + \{I_{ab}\}} \qquad \frac{\square}{\{I_{ab}\} * (c) \{I_{abc}\}}}{\{I_{ab\overline{c}}\} + \sim * (c) \{I_{abc}\}}$$
$$\frac{\{i \in [a, b]\} * (1-c) : + \sim * (c) \{o \in [a, b]\}}{\{i \in [a, b]\}}$$

with:

$$\begin{array}{ll} I_{ab}(x) &\equiv x \in [a,b] \\ I_{abc}(x) &\equiv x \in [a * c, b * c] \\ I_{ab\overline{c}}(x) &\equiv x \in [a * (1-c), b * (1-c)] \end{array}$$

Stability of Smooth

Three main VC in the proof:

(* (1 - c) * i \in [(1 - c) * a, (1 - c) * b] *)
by rewrite ?ler_wpmul2r ?ler_subr_addr ?add0r.

have Ha: a = a * c + a * (1 - c)by rewrite -mulrDr addrC addrNK mulr1. have Hb: b = b * c + b * (1 - c)by rewrite -mulrDr addrC addrNK mulr1. by rewrite Ha Hb !ler_add.

(* c * i \in [c * a, c * b] *)
by rewrite ?ler_wpmul2r.

We pushed the VCs to Why3 with success. Interval technique ready to go into the main compiler.

Stability Proof



One Step Beyond

Extending the logic

Allow predicates to refer to windows.

$$\varphi(i) \equiv \{i/i_{\Box} = 0.8\}$$

where i_{\Box} is the sample produced in the execution step.

Linear System Theory

Consider the following subset of Faust:

- *(c) scaling by c
- + addition
- : composition
- \sim addition

Then every Faust program is LTI. Very related to [Bonchi et al. 2015] A consequence of that is that every program can be viewed

as a polynomial.

Two Poles IIR Filter

```
twopole = fir : + ~ feedback
where
fir(x) = (x - x'') * g * (1-RR) / 2;
feedback(v) = 2*R*cos(T) * v - RR * v';
....
```

Two Poles IIR Filter

Get and verify its transfer function:

$$H(z) = \frac{1 - z^{-2}}{1 - 2R\cos(\Theta_c)z^{-1} + R^2 z^{-2}}$$

Ongoing: Frequency Domain Analysis

Recall the Fourier Matrix:

$$W = 1/\sqrt{N} \begin{bmatrix} 1 & 1 & 1 & \cdots & 1 \\ 1 & \omega & \omega^2 & \cdots & \omega^{N-1} \\ 1 & \omega^2 & \omega^4 & \cdots & \omega^{2(N-1)} \\ \vdots & \vdots & \vdots & \ddots & \vdots \\ 1 & \omega^{N-1} & \omega^{2(N-1)} & \cdots & \omega^{(N-1)(N-1)} \end{bmatrix}$$

or:

$$\boldsymbol{W} = \left(\frac{\omega^{jk}}{\sqrt{N}}\right)_{j,k=0,\ldots,(N-1)}$$

where ω the nth-root of the unity. Then the DFT can be expressed as:

$$X = Wx$$

Fourier Properties Formally

Linearity, shifting and scaling follow from lemmas already in the MathComp linear algebra library! Parseval's theorem is work in progress:

$$\sum_{n=0}^{N-1} |x_n|^2 = \sum_{n=0}^{N-1} |X_n|^2$$

Transfer Functions

- We can use a similar approach for the certification of transfer functions.
- We use the finite Z-transform, plus some caveats, mainly about the bounds.

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- C.f: Algebraic Signal Processing [Puesel, Moura]

Paper with our adventures coming end of month.

Conclusions

- It was an interesting exercise; we learned a lot!
- The full Faust language is basically done.
- So far verification has been about math verification.
- Floating point issues ignored...
- Help from audio people. What are important things to certify?
- Non-Linear systems.
- We are investigating a different approaches to certification beyond program logics.
- Verified FFT/DSP computation. Trying CoqEAL.
- Improving the language for spectral processing.
- Non-linear Wave Filter, Scattered Delays Networks.