

A model for interactive media authoring

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Abstract: A single paragraph of about 200 words maximum. For research articles, abstracts should give a pertinent overview of the work. We strongly encourage authors to use the following style of structured abstracts, but without headings: 1) Background: Place the question addressed in a broad context and highlight the purpose of the study; 2) Methods: Describe briefly the main methods or treatments applied; 3) Results: Summarize the article's main findings; and 4) Conclusion: Indicate the main conclusions or interpretations. The abstract should be an objective representation of the article, it must not contain results which are not presented and substantiated in the main text and should not exaggerate the main conclusions.

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1. Introduction

Many music software fit in one of three categories: sequencers, patchers, and textual programming environments. Sequencers are used to describe temporal behaviours: an audio clip plays after another, while an automation curve changes an audio filter. Patchers are more commonly used to describe invariants: for instance specific audio filters, or compositional patterns.

We propose in this paper a method that combines the sequencer and the patcher paradigm in a live system.

The general approach is as follows: we first introduce a minimal model of the data we are operating on: namely, remote software or hardware such as OSC peripherals and sound cards. Then, two structures are presented: the first is a temporal structure, which allows to position events and processes relatively to each other, hierarchically, and in a timely fashion. The second is a graph structure akin to dataflows. This graph uses special connection types to take into account the fact that nodes of the graph might not always be active at the same time. Both structures are then combined: the state of the temporal processes is bound to the dataflow nodes. This combination is then expanded with specific implicit cases that are relevant in computer music workflows. These cases are described using structures wrapping the temporal and dataflow graphs.

We compare the various models in the context of music creation: what entails using only the temporal structure, only the graph structure, and the combination of both.

The latter model is shown to have enough expressive power to allow for recreation of common audio software logic within it: for instance traditional or looping audio sequencers. Additionally, its

use is presented in sample compositions: the first one is an example of audio editing, the second an interactive musical installation.

1.1. State of the art

There is a long-standing interest in the handling of time in programming languages, which is intrinsically linked to how the language handles dynamicity.

PEARL90[?] ¹ provides temporal primitives allowing for instance to perform loops at a given rate for a given amount of time. More recently, Céu has been introduced as a synchronous language with temporal operators, and applications to multimedia[?].

OpenMusic is a visual environment which allows to write music by functional composition. It has been recently extended with timed sequences allowing to specify evolutions of parameters in time[?].

Likewise, the Bach library for Max [?] allows to define temporal variations of parameters during the playing of a note by with the mechanism of slots. The processes controlled by such parameters are then available to use in the Max patch.

The Max for Live extension to Ableton Live allows to embed Max patches in the Ableton Live sequencer. Through the API provided, one can control the execution of various elements of the sequencer in Max; automations in Live can also be used to send data to Max patches at a given time.

A method for dynamic patching of Max abstractions based on CommonLisp has been proposed by Thomas Hummel[?] to reduce resource usage by enabling and disabling sub-patches at different points in the execution of a program. This has the advantage of saving computing power for the active elements of the score.

Dataflows and especially synchronous dataflows have seen tremendous usage in the music and signal processing community. A list of patterns commonly used when developing dataflow-based music software is presented in [?]. Formal semantics are given in [?]. Specific implementation aspects of dataflow systems are discussed in the Handbook of Signal Processing Systems[?].

Dynamicity in dataflows is generally separated in two independent aspects: dynamicity of the data, and of the topology. The first relates to the variability on the streams of tokens, while the second is about changes to the structure of the graph. Boolean parametric dataflows[?] have been proposed to solve dynamicity of topology, by introducing conditionals at the edges.

base: max, pd, séquenceurs: cubase/protools , live/bitwig...

openmusic

antescofo

inscore

1.2. Context of this research

This paper follows existing research on interactive scores, as part of the i-score project. Previous research focused on operational semantics for interactive scores, based on time automatas[?] or Petri nets[?], mainly for software verification purposes. In contrast, we give here domain-centered functional semantics which models the current C++ implementation of the software.

We first define the temporal model, then extend it with a distinct data model which reads and produces the various inputs & outputs of the system. Then, we introduce implicit operations and defaults in the context of a GUI software to create, modify, and playback such scores. These operations allow to simplify the usage of the paradigm for composers. Real-world examples are provided and discussed.

[?]

¹ Not to be mistaken with the Perl language commonly used for text processing

2. Orchestrated data

We first define the data we operate on. External devices are modeled as a tree of optional parameters.

Value parameters can have values of common data types such as integer, float, etc. Audio parameters are arrays that contain either the current input audio buffers of the sound card or the buffers that will be written to the sound card's output.

The tree of nodes is akin to the methods and containers described in the OSC specification.

$$\mathbf{Value} = \text{Float} \mid \text{Int} \mid \text{Bool} \mid \text{String} \mid \dots$$

$$\mathbf{ValueParameter} = \text{Value} \times \text{Protocol}$$

$$\mathbf{AudioParameter} = \text{Float}[] \times \text{Protocol}$$

$$\mathbf{Parameter} = \text{ValueParameter} \mid \text{AudioParameter}$$

$$\mathbf{Node} = \text{String} \times \text{Maybe Parameter} \times \text{Node}[]$$

Parameters and nodes bear additional metadata which is not relevant to describe here: textual description, tags, etc.

The parameters's associated values match the state of an external device: synthesizer, etc. Multiple protocols are implemented to allow this: for instance OSC, MIDI, etc.

We define two core operations on parameters:

$$\mathbf{pull} : \text{Parameter} \rightarrow \text{Parameter}$$

$$(v, p) \mapsto (v', p) \text{ where } v' \text{ is the last known value in the remote device}$$

$$\mathbf{push} : \text{Parameter} \times \text{Value} \rightarrow \text{Parameter}$$

$$(v, p), v' \mapsto (v', p) \text{ and } v' \text{ is sent to the remote device}$$

3. Temporal model

The temporal model is twofold: it is a hierarchical tree of processes, whose durations and execution times are directed by intervals. The beginning and end of the intervals is subjected to various conditions that will be presented ; these conditions allow for various interactive behaviours. Processes are any kind of relevant artistic process: automations, notes sequences, sound files. In particular, specific dispositions of intervals and conditions are implemented as processes themselves, scenario and loop, which allows for hierarchy.

We note: TC for temporal conditions, IC for instantaneous conditions, I for intervals, P for processes. First, these elements are defined, then the semantics imposed on them by the scenario and the loop are presented. These semantics allow both serial and parallel execution of musical processes.

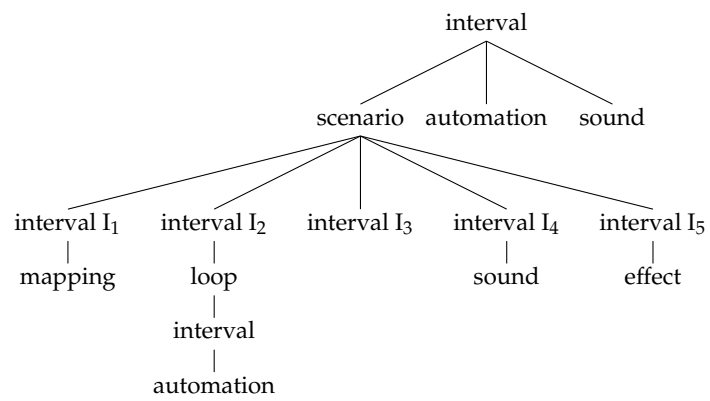
3.1. Data types

3.1.1. Conditions and expressions

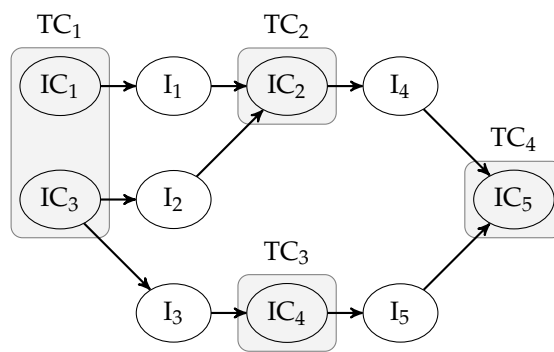
We first define the conditional operations we want to be able to express. We restrain ourselves to simple propositional logic operands: **and**, **or**, **not**.

Expressions operate on addresses and values of the device tree presented in 2.

Formally, expressions are defined as a tree: Let **Comparator** be an identifier for standard value comparison operations: $<, \leq, >, \geq, =, \neq$ and **Operator** standard logical operators **and** & **or**.



(a) Hierarchical tree



(b) Temporal DAG

$\mathbf{Atom} : (\text{Parameter} \mid \text{Value}) \times (\text{Parameter} \mid \text{Value}) \times \text{Comparator}$
 $\mathbf{Negation} : \text{Expression}$
 $\mathbf{Composition} : \text{Expression} \times \text{Expression} \times \text{Operator}$
 $\mathbf{Impulse} : \text{Parameter} \times \text{Bool}$
 $\mathbf{Expression} : \text{Atom} \mid \text{Negation} \mid \text{Composition} \mid \text{Impulse}$

Two operations are defined on expressions and the data types that compose them:

- **update** : $\text{Expression} \rightarrow \text{Expression}$. Used to reset any internal state and query up-to-date values for the expressions. For instance, **update** on an **Atom** fetches if possible new values for the parameters, why may include network requests.

Precisely:

$$\left\{ \begin{array}{l}
 \text{update} : \text{Composition} \rightarrow \text{Composition} \\
 \quad (e_1, e_2, o) \mapsto (\text{update } e_1, \text{update } e_2, o) \\
 \text{update} : \text{Negation} \rightarrow \text{Negation} \\
 \quad e_1 \mapsto \text{update } e_1 \\
 \text{update} : \text{Atom} \rightarrow \text{Atom} \\
 \quad \left\{ \begin{array}{l}
 (\text{parameter } p_1, \text{parameter } p_2, o) \mapsto (\text{pull } p_1, \text{pull } p_2, o) \\
 (\text{parameter } p_1, \text{value } v_2, o) \mapsto (\text{pull } p_1, v_2, o) \\
 \dots
 \end{array} \right. \\
 \text{update} : \text{Impulse} \rightarrow \text{Impulse} \\
 \quad (p, b) \mapsto (p, \text{false})
 \end{array} \right.$$

- **evaluate** : $\text{Expression} \rightarrow \text{Bool}$. Performs the actual logical expression evaluation, according to the expected logical rules.
- An atom is a comparison between two parameters, a parameter and a value, or two values.
- Negations and compositions are the traditional predicate logic building blocks.
- We introduce a specific operator, “impulse”, which allows to decide whether a message was received in a given time span.

3.1.2. Temporal processes

Temporal processes are executed by intervals at each tick. The actual processes will be defined in sections , 3.2, 3.3, 4.

A process is a type P associated with the following operations:

$\mathbf{start} : P \rightarrow P$
 $\mathbf{stop} : P \rightarrow P$
 $\mathbf{tick} : P * \mathbb{Z} * \mathbb{Z} * \mathbb{R} \rightarrow P$
 $\mathbf{Process} : \text{Scenario} \mid \text{Loop} \mid \dots$

3.1.3. Interval

We want to be able to express the passing of time, for a given duration. This duration may or may not be finite.

A duration is defined as a positive integer. An interval is at its core a set of durations: a min, an optional max, and the current position. The lack of max means infinity. An interval is said to be fixed when its min equals its max. It may be enabled or disabled.

Status = Waiting | Pending | Happened | Disposed

Interval = Duration \times Maybe Duration \times Duration \times Status

The time scale is not specified by the system: for instance, when working with audio data it may be better to use the audio sample as a base unit of time. But many applications don't use the audio rate: when working purely with visuals it may be better to use the screen refresh rate as time base in order not to waste computer resources and energy.

3.1.4. Instantaneous condition

Then, we want to be able to enable or disable events and intervals according to a condition, given in the expression language seen in ???. An instantaneous condition is defined as follows:

Condition = Expression \times Interval[] \times Interval[] \times Status

It is preceded and followed by a set of intervals.

Expressions are disabled either when they are false or when they are preceded by a non-null number of intervals, all of them already disabled through other conditions. This propagates recursively to the following intervals and conditions.

3.1.5. Temporal condition

A temporal condition is used to synchronize starts and ends of intervals, while allowing to implement behaviours such as : "start the chorus when the fader is at 0".

Temporal conditions carry instantaneous conditions, which will be evaluated at the moment where the temporal condition becomes true.

A note on asynchronicity

We consider here a synchronous version of the temporal conditions, which assumes that external values would not variate at a greater rate than the tick rate of the system; in practice this is generally not the case. Hence, the actual implementation of temporal conditions supports asynchronous updates of the condition value according to received network messages rather than the tick rate. This means that the execution will not miss cases where a condition would become false, true, then false again after the beginning and before the end of a tick.

3.1.6. Operations

```

type process =
  NodeProcess of nodeProcess | Scenario of scenario | Loop of loop
and interval = {
  minDuration: duration;
  maxDuration : duration option;
  nominalDuration : duration;
  itvStatus: status;
  processes: process list
}
and condition = {

```

```

155 condExpr: expression;
156 previousItv: interval list;
157 nextItv: interval list;
158 status: status;
159 }
160 and temporalCondition = {
161   syncExpr: expression;
162   conds: condition list
163 }
164 and scenario = {
165   intervals: interval list ;
166   triggers: temporalCondition list;
167 }
168 and loop = {
169   pattern: interval;
170   startTrig: temporalCondition;
171   endTrig: temporalCondition;
172 };;

173 add_process interval proc: interval * proc -> interval
174   (t1, t2, p, t3) -> (t1, t2, proc::p, t3)

175 add_event tc ic: TemporalCond * InstCond -> TemporalCond
176   (... , ics , ...) -> (... , ic::ics , ...)

177   exécution :
178   interval:
179   On retourne une fonction ... graph_fun va être appliquée au graph. Faut-il avoir une liste explicite
180   ou bien juste passer le graph et demander à chaque fonction d'appliquer ses fonctions enfants ? Le
181   second fait plus fonctionnel mais le premier laisse moins de marge d'erreur (DRY)

182 get_node graph node_id -> node
183 update_node graph node_id node -> graph
184
185 graph_fun: graph -> graph ; va transformer un noeud du graphe d'une maniere donnee
186
187 tuple_first tpls: retourne les premiers elements d'une liste de paires
188 tuple_second tpls: retourne les premiers elements d'une liste de paires
189
190 tick: itv , count , offset : interval * duration * duration -> interval * graph_fun[]
191   ((... , nom, t, pos, procs), new_date) -> (
192     let procs = map procs (state _ t offset) in
193     (... , t + count, t + count / nom, tuple_first procs ),
194     fun (node_date,node_offset) -> (t+count, offset) :: tuple_second procs)

195   processes:
196   state: process * t -> process * graph_fun
197
198   described for each process (polymorphic)

```

199 3.2. Temporal graph: scenario

200 3.2.1. Creational operations

201 `add_interval sc itv sev eev`

202 `add_sync sc`

203 3.2.2. Execution operations

204 `process_event :`

205 `make_happen :`

206 `make_dispose :`

207 `scenario_state : scenario → scenario * state`

208 3.3. Loop

209 `Pbq`: not introducing cycles in the temporal graph

210 `process_event :`

211 `make_happen :`

212 `make_dispose :`

213 `loop_state :`

214 4. Data model

215 `=> set date => set offset pour offset audio (p-ê pas nécessaire si on fait comme LStream)`

216 5. Data graph

217 `Questions: * node ordering * port definitions`

218

219 `(* ports *)`

220 `type edgeType = Glutton | Strict | Delayed ;;`

221 `type edge = { edgeId: int; source: int; sink: int; edgeType: edgeType; }`

222 `and audioPort = { audioPortId: int; audioPortAddr: audioParameter option; audioEdges: e`

223 `and valuePort = { valuePortId: int; valuePortAddr: valueParameter option; valueEdges: e`

224 `and port = AudioPort of audioPort | ValuePort of valuePort`

225 `;;`

226 5.1. Structures

227

228 `(* curves *)`

229 `type curve = (float * float) list ;;`

230 `let value_at curve x = 0.0;;`

231


```

232 (* some processes *)
233 type automation = valuePort * curve;;
234 type mapping = valuePort * valuePort * curve;;
235 type sound = audioPort * float array array;;
236 type passthrough = audioPort * valuePort * audioPort * valuePort;;
237
238 type dataNode = Automation of automation | Mapping of mapping | Sound of sound | Passth
239 type grNode = { nodeId: int; data: dataNode; enabled: bool; date: duration; position: p
240
241
242 type graph = { nodes: grNode list ; edges: edge list; };;
243
244 let next_id lst f = 1 + (List.fold_left max 0 (List.map f lst));;
245 let next_node_id lst = next_id lst (fun n -> n.nodeId);;
246 let next_edge_id lst = next_id lst (fun n -> n.edgeId);;
247
248 let create_audio_port = { audioPortId = 0; audioPortAddr = None; audioEdges = []; } ;;
249 let create_value_port = { valuePortId = 0; valuePortAddr = None; valueEdges = []; } ;;
250 let test_edge = { edgeId = 33; source = 4; sink = 5; edgeType = Glutton; };;
251 let some_sound_data = Array.make 2 (Array.make 8 0.);;
252 let some_sound = Sound (create_audio_port, some_sound_data);;
253
254 let some_passthrough = Passthrough ( create_audio_port, create_value_port, create_audio
255
256 (* test *)
257 let test_node_1 = { nodeId = 1; data = some_sound; enabled = false; date = 0; position
258 let test_node_2 = { nodeId = 34; data = some_sound; enabled = false; date = 0; position
259 next_node_id [ test_node_1; test_node_2 ] ;;
260
261 let create_graph = { nodes = []; edges = [] };;
262 let add_node gr nodeDat =
263 let new_id = next_node_id gr.nodes in
264 let newNodeDat = match nodeDat with
265 | Automation a -> nodeDat
266 | Mapping m -> nodeDat
267 | Sound s -> nodeDat
268 | Passthrough p -> nodeDat
269 in
270 let new_node = { nodeId = new_id; data = newNodeDat; enabled = false; date = 0; positio
271 (new_node, {nodes = new_node::gr.nodes; edges = gr.edges})
272 ;;
273 let add_edge gr src snk t =
274 let new_id = next_edge_id gr.edges in
275 let new_edge = { edgeId = new_id; source = src; sink = snk; edgeType = t } in
276 (new_edge, { nodes = gr.nodes; edges = new_edge::gr.edges })
277 ;;
278
279 (* test *)
280 let test_g = create_graph;;
281 let (snd1, test_g) = add_node test_g some_sound;;

```

```

282 let (snd2, test_g) = add_node test_g some_sound;;
283 let (p1, test_g) = add_node test_g some_passthrough;;
284
285 (* let (e1, test_g) = add_edge snd1. *)
286
287
288 type nodeProcess = {
289   node: int;
290   curTime: duration;
291   curOffset: duration;
292   curPos: position;
293 };;
294
295
296
297
298
299
300 node: enabled * executed * time * position * inlets * outlets * priority
301 add_node graph
302 connect graph node node edge
303 enable graph node
304 disable graph node

```

305 5.2. Operations

306 Input mix on each port

```

307 copy_from_global
308 copy_from_local
309 init_node
310 teardown_node

```

311 5.3. Tick description

```

312     General flow:
313     disable strict nodes
314     sort remaining nodes according to the custom order chosen (default, temporal, custom)
315     priority: * explicit cables * local or global address
316     do a tick:

```

```

317 let clear_outputs n =
318   (_, ..., (map n.outputs (match p with
319     | value -> clear value
320     | audio -> clear audio
321   )))
322
323 let pull_port p : port -> port
324
325 init_value : port -> value
326 let init_value value_port =
327   if !empty value_port.cables

```

```

328     mix (pull_port value_port.cables)
329     pull value_port.address
330
331
332   let init_node g n =
333     (_, (map n.inputs (match p with
334       | value -> pull value
335     )), ...)
336   exec_node :
337     in
338     let copy_inputs n =
339     let init_node n =
340       copy_inputs clear_outputs n
341     in
342
343     new_node, new_local_state = exec_node init_node n
344     replace g n new_node
345
346   tick_graph :
347     while: ! empty nodes

```

348 5.4. Data nodes

349 5.4.1. Passthrough

350 -> used for scenario and interval -> mixing at the input

351 5.4.2. Automation

352 Curves

353 start point + set of (segment * breakpoint)
 354 curve + message output port $x \in [0;1]$ -> in the nominal duration of the parent time interval.
 355 state_autom :

356 5.4.3. Mapping

357 message input port + curve + message output port
 358 state_mapping :

359 5.4.4. JavaScript

360 n message input port + curve + n message output port
 361 state_js :

362 5.4.5. Piano Roll

363 notes + midi output port
 364 state_midi :

Figure 2. General data flow for a tick

365 5.4.6. Sound file

366 sound data + midi output port

367 state_sndfile :

368 5.4.7. Buffer

369 Used to keep audio input in memory

370 Why isn't the delay cable not enough ? can't go backwards. pb: pauser au milieu: coupure. cas
371 dans les boucles: on réécrit par dessus (buffer vidé sur start).

372 state_buffer :

373 6. Combined model

374 -> on ajoute node aux tc

375 -> nodeprocess fait le lien entre graphnode et time process, permet de faire l'activation et
376 l'écoulement du temps

377 -> offset nécessaire pour tc pour gérer l'audio (mais pourrait être ajouté dans le modèle de base.
378 Ou bien passer une paire de pointeurs.. ?)

379 6.1. Combined tick

380 Exécution complète d'un tick: Copy audio buffers and input data, execute the temporal tick,
381 execute the graph tick, copy the output audio buffer and apply the produced state by pushing the
382 values.

383 Pour être propre, il faudrait faire un "pull" général au début...

384 7. Proposed sequencer behaviour

385 Conditions et cie: The most common case for an expression is to be true.

386 UI: création automatique de liens implicites des enfants vers les parents => "cable créé par défaut"
387 quand on rajoute un processus dont on marque l'entrée

388 => pour toute contrainte, pour tout scénario, créer noeud qui fait le mixage => création d'objets
389 récursivement, etc

390 - Problème des states dans scénario ? => states du scénario: comment interviennent-ils ? faire un
391 scénario fantôme

392 - Mettre l'accent sur la recreation de la sémantique de i-score à partir du graphe: => messages:
393 actuellement "peu" typés ; rajouter type de l'unité ?

394 => pbq du multicanal: pour l'instant non traitée, on ne gère que les cas mono / stereo pour
395 le upmix / downmix Choix pour multicanal: faire comme jamoma avec objets tilde => sliders et
396 dispatching de canaux ? => cables: rubberband ? il faut mettre un rubberband dès qu'on a une entrée
397 et une sortie qui n'ont pas la même vitesse relative. Dire que pour les automatisations ça interpole de
398 manière naturelle avec le ralentissement et l'accélération (on sépare vitesse et granularité)

399 - Dire qu'on pourrait affiner en combinant plus précisément les "sous-ticks" temporels et de
400 données pour que par exemple la production d'un état dans un scénario entraîne une condition dans
401 un autre scénario

8. Applications and examples

8.1. Reconstructing existing paradigms

In this part we give example of reconstruction of standard audio software behaviours with the given model.

8.1.1. Audio sequencer

Notable software in this category includes Steinberg Cubase, Avid Pro Tools, ...

The common metaphor for audio sequencers is the track, inspired from mixing desks and tape recorders. We will take the example of audio and midi tracks. Such an audio sequencer can be modeled by :

- A root: an infinite interval.
- This interval contains two processes: a scenario and an effect bus. The sound output of the scenario goes to the input of the effect bus.
- The scenario contains the actual tracks.
- These tracks are also modeled by infinite constraints.

We divide the tracks in two categories. Audio tracks are built with :

- A scenario with a single sequence of intervals, some of which may bear sound file processes and others being empty.
- An effect bus process. The output of the scenario goes to the input of the effect bus. Generally, this effect bus would end by channel operations such as panning and volume adjustment, in a similar fashion to mixing desks.

Midi tracks are built with :

- A scenario with a single sequence of intervals, some of which may bear MIDI notes processes and others being empty.
- An instrument process, which takes MIDI data and outputs sound.
- Like before, an effect bus applied to the instrument's output.

This can easily be extended with further features: sends, automations, etc.

8.1.2. Looping audio sequencer

More recently, a different kind of sequencer has emerged: the looping, non-linear sequencer. The prime exemple of this is Ableton Live. We give the example for a simplified model of live-looping without quantization.

These sequencers are also organized in tracks ; however, within a track, the musician can choose a single loop that is currently playing, and regularly switch the current loop.

Hence, the general organization stays the same than for the audio sequencer: most importantly, the way effect buses are applied does not change.

- Each clip of a track is given an index.
- Each track also has a parameter which is the next clip to play, `next_clip`. These parameters can be introduced as variables in the device tree.
- We replace the scenarios containing the actual sound files or midi notes by loop processes.
- The loops processes are defined with an ending temporal condition.
- Inside the loop pattern, there is a single scenario process. This scenario process has a set of parallel intervals, each with one sound file. Every interval begins with an instantaneous condition that compares the `next_clip` parameter to the current clip's index. Hence, at most one clip is playing at the same time in each track. If the `next_clip` does not change the track keeps looping on the sound file.

Extension: par ex. dans une boucle on peut mettre un autre scénario. Pb : tic qui manque. On peut y remédier en exécutant le trigger "en avance".

8.1.3. Patcher

8.2. Musical examples

8.2.1. Audio compositing

-> on utilise un scénario qui lit des parties d'une entrée son dans différents bus d'effets. L'effet peut se déclencher en retard.

Org:

Intervalle racine

Process 1: Audio input Process 2 : Scenario -> Trois itv ; entrée reliée strict à sortie de audio input ; sortie dans parent

Process 3 : FX Process 4 : scenario Audio Input -> itv 1,2,3 -> scenario -> fx -> scenario

8.2.2. Musical carousel

We present here a real-world interactive music example: the musical carousel. Each seat on the carousel has different instrument-like input devices: reactive pads, motion sensors, etc. A run in the carousel generally operates as follows:

- The first few seconds, the rules of the carousel are explained to the participants.
- The song starts: the passengers can start interacting with their instruments. An overall music is generated from their interpretation. Played notes stay in predefined scales which may vary over time; pre-recorded parts can also be layered on top. The overall song structure can vary according to the intensity of the played music: for instance, if everyone plays *piano*, different instruments may become available in the next section of the song, a part may be shorter or longer, etc. Such variations are written by the composer for each song.
- At the end of the song, the participants hear a summarized version of the song they just played. This version also has additional corrections and adjustments applied algorithmically.

- Réutilisation des données d'entrée: scores sur certaines parties ; réutilisation de certaines notes et des pics d'intensité -> nécessite d/dx - Gammes: filtrage global du MIDI In

8.3. Notes on implementation

=> "third gen" audio sequencer. first gen: cubase, etc second gen: non-linear: ableton, bitwig third gen: entirely interactive: i-score, iannix. what else ?
reproductibilité: code source dispo

9. Evaluation and Discussion

Enforcing graph constraints: mostly done through UI. For instance: ic are created on tc, etc. No "going back" which would break DAG-ness.

Faire parenthèse sur domain driven design sur logiciels de musique qui fournit de meilleurs résultats que application directe de modèles existants (petri, etc). Peut-être donner un méta-modèle qui correspond à nos structures ?

Dire pourquoi un tic est introduit lors d'une interaction (notamment, permet de ne pas avoir de "boucle infinie" si on a une boucle de durée 0 avec deux triggers vrais) ; est aussi plus cohérent pour les utilisateurs pour qui une interaction doit être manifeste.

Avantage: manipulation uniforme des processus, que ce soit des automatisations, des groupes, des fichiers sons, etc.

10. Conclusion

missing: quantification

missing: sound speed

Supplementary Materials: The following are available online at www.mdpi.com/link, Figure S1: title, Table S1: title, Video S1: title.

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