

‘Continuous-Time’ Synthesis and the Context of Timelab’

March 14, 2014

1 Introduction

1.1 why timelab?

1.1.1 issues with existing systems

Music-N/CSound (Max Mathews and John ffitch, 1957-present) A ‘classic’ programming environment, stemming from the pioneering work of Max Mathews, it is widely used, and highly developed.

- rigid structure
- difficult to interact with / compile time
- obtuse representation of scores and synths / hard to read
- score/instrument layout insists on antique perception of music composition/creation – a notion which in many ways goes against the grain of analog electronic music practices

Max/Msp and Pd (David Zicarelli and Miller Puckette, 1987-present) An almost ubiquitous and highly interactive paradigm, these programs the are industry standards (commercial and open source respectively) for experimental audio algorithm design.

- patch chords and guis!!!
- difficulty of iterating loops
- flow diagram structure (borrowing from analog synthesizers) also insists on a way of thinking about music composition – denying the user a ‘code-ish’ mindset

SuperCollider (James McCartney, 1996-present) SC provides the interactivity of Pd with the ‘codiness’ of a text based programming environment such as CSound.

- server/client structure has benefits but designates a .2s default latency between server time and client time
- non-standard editor and gui environments (not necessarily a drawback)

ChucK (Ge Wang and Perry Cook) ChucK is another text-based audio programming environment that emphasizes a live-coding ethos.

- admittedly emphasizes ease of programming over performance (ChucK is slow)
- unintuitive approach to time by treating ‘now’ as a variable that we manipulate by hand: <http://en.wikipedia.org/wiki/Chuck>

1.1.2 Fall 2012 206

1.1.3 Control timing

All of the above mentioned systems have specific solutions to problems of rectifying the timing between control input and precisely when control is enacted in the DSP engine. Cf. my poster from last year’s ICMC (TIMELAB: YET, YET ANOTHER REAL-TIME AUDIO PROGRAMMING SYSTEM – not yet available online) for a quick and dirty on this subject.

Also of relevance are these talks by Miller Puckette and James McCartney: <http://repmus.ircam.fr/mutant/rtmseminars>

- Puckette talk: 9:30 - 15:00 : DAGs and mutual exclusion, determinism
- McCartney talk: 29:00 - 37:45 : control rate issues

1.1.4 embedded audio programming

Just the other day (Friday, March 14) someone posted to the Pd-list asking if it was possible to crunch a Pd patch down into a guitar pedal (the answer was ‘no, not really’). Clearly, there is a market for throwing these kinds of experimental algorithms into a guitar pedal packages. One solution is to utilize micro computers such as the Raspberry Pi or UDOO. Another is to have an audio programming API in C with a build environment that can target truly embeddable hardware, such as ST’s discovery board featuring the ARM Cortex processor (or any of the open source solutions built around these chips).

2 Continuous-Time Synthesis

2.1 0-time delay lines

2.2 virtual analog

2.2.1 LTspice

2.2.2 wave-digital filters

2.2.3 moog filter example

2.3 new models for new sounds

2.3.1 noisy oscillators

examples – applications (lfo)

2.3.2 non-linear filters

???

2.3.3 a new way to view synthesis

2.4 What's Next?

3 Timelab

3.1 Brief Overview

3.2 Examples