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MapLooper: Live-looping of distributed gesture-to-sound mappings

Christian Frisson[1], Mathias Bredholt[1], Joseph Malloch[2], Marcelo M. Wanderley[1]

[1]Input Devices and Music Interaction Laboratory (IDMIL), Centre for Interdisciplinary Research in Music Media and Technology (CIRMMT), McGill University,
[2]Graphics and Experiential Media (GEM) lab, Dalhousie University

ABSTRACT

This paper presents the development of MapLooper: a live-looping system for gesture-to-sound mappings. We first reviewed loop-based Digital Musical Instruments (DMIs). We then developed a connectivity infrastructure for wireless embedded musical instruments with distributed mapping and synchronization. We evaluated our infrastructure in the context of the real-time constraints of music performance. We measured a round-trip latency of 4.81 ms when mapping signals at 100 Hz with embedded *libmapper* and an average inter-onset delay of 3.03 ms for synchronizing with Ableton Link. On top of this infrastructure, we developed *MapLooper*: a live-looping tool with 2 example musical applications: a harp synthesizer with SuperCollider and embedded source-filter synthesis with FAUST on ESP32. Our system is based on a novel approach to mapping, extrapolating from using FIR and IIR filters on gestural data to using delay-lines as part of the mapping of DMIs. Our system features rhythmic time quantization and a flexible loop manipulation system for creative musical exploration. We open-source all of our components.

Author Keywords

Digital Musical Instrument, mapping, looping, synchronization, embedded computing

CCS Concepts

Applied computing → **Media arts; Hardware** → *Sensor devices and platforms; Sound-based input / output.*

Introduction

Composers Pauline Oliveros and Terry Riley explored technology-driven repetition in music in the 1950s through pioneering experiments with tape loop techniques and tape delay/feedback systems [1]. Their system, *Time Lag Accumulator*, worked by stringing tape between two tape recorders and feeding the signal from the second machine back to the first, mixing incoming sound with the tape's previously recorded sound. Later, digital looping devices re-implemented this concept. Digital memory replaced magnetic tape, and digital loopers are now available in much smaller form factors than magnetic tape recorders.

A Digital Musical Instrument (DMI) consists of a gestural interface and a sound generation [2]. The gestural interface and sound generator are separate units related

by mapping strategies. Hunt et al. demonstrated [3] that different mappings can completely change an instrument's behavior.

Mappings have been employed in synthesis engines [4], physical models [5], or audio effects [6]. In these contexts, mappings facilitate skill-based performance, characterized by rapid, coordinated movements in response to continuous signals [7]. This type of performance often involves instruments with a high level of mapping *transparency*, where the link between a performer's gesture and the resulting sound is clear to both audience and performer, correlating with instrument expressiveness [8]. Musicians seeking the aesthetics of accurate and precise timing typically require a high skill level, while existing tools for creating loop-based music such as music sequencers, samplers, and loopers offer beginners a low "entry fee" [9]. However, the control mapping of these tools is often opaque and difficult for the audience to understand. In this work, we explore mapping in the context of loop-based music performance with the goal of creating instruments with a low entry fee and high mapping transparency.

In this paper, we first review several looping tools and list our design requirements. We then describe our mapping and synchronization platform for embedded devices, and validate our approach through the gesture-to-sound looping tool *MapLooper* and two example synthesis applications. We finish by discussing perspectives beyond our work.

Related work

We review several looping tools involving gesture-to-sound mappings grouped into two main categories: a) audio stream loopers, b) control data stream loopers.

Audio stream loopers

Audio stream loopers have become popular in the form of commercial live-looping pedals. These devices usually have user interfaces with buttons and knobs for controlling recording and playback states, loop length, and volume of loop layers. Loop controls can also be controlled gesturally, giving the performer the possibility to perform with gestures and body movements.

SoundCatcher

SoundCatcher [10] (Image 1) is a live-looping system with a mid-air gestural control interface. The distance between the performer's hands is mapped to the loop length

and vibrotactile feedback. *SoundCatcher* is an example of the usage of an explicit mapping strategy for the control of live-looping.
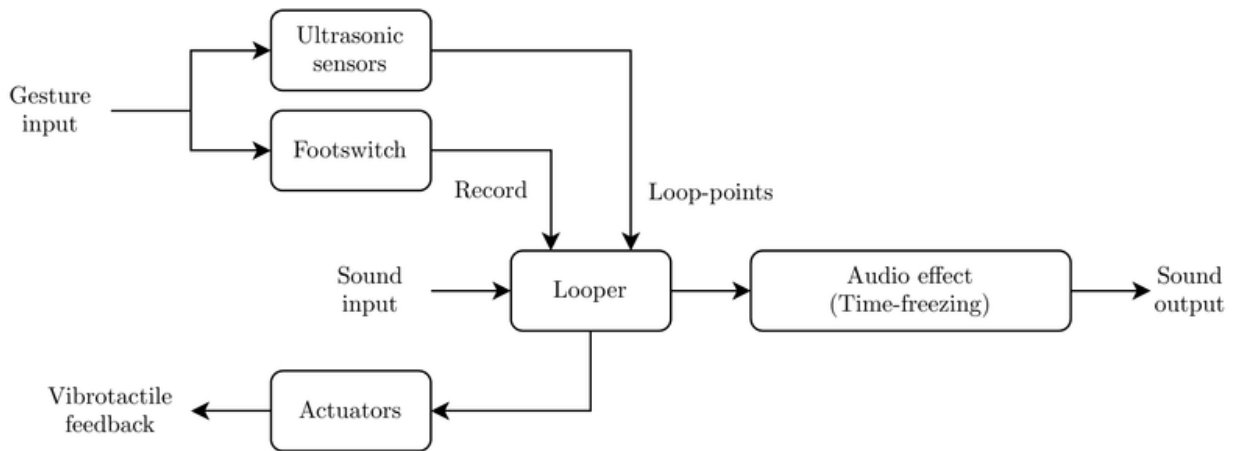

Image 1

Gesture-to-sound interface of *SoundCatcher*. The performer is holding the actuators, and the ultrasonic sensors are mounted to a microphone stand.

SoundGrasp

SoundGrasp [11] (Image 2) features a mid-air gestural control interface with a glove controlling the recording/playback state and parameters for reverb and echo effects. Postures are classified into a vocabulary of control commands such as record/play/stop. *SoundGrasp* is an example of using machine learning as a mapping strategy for the control of live-looping.
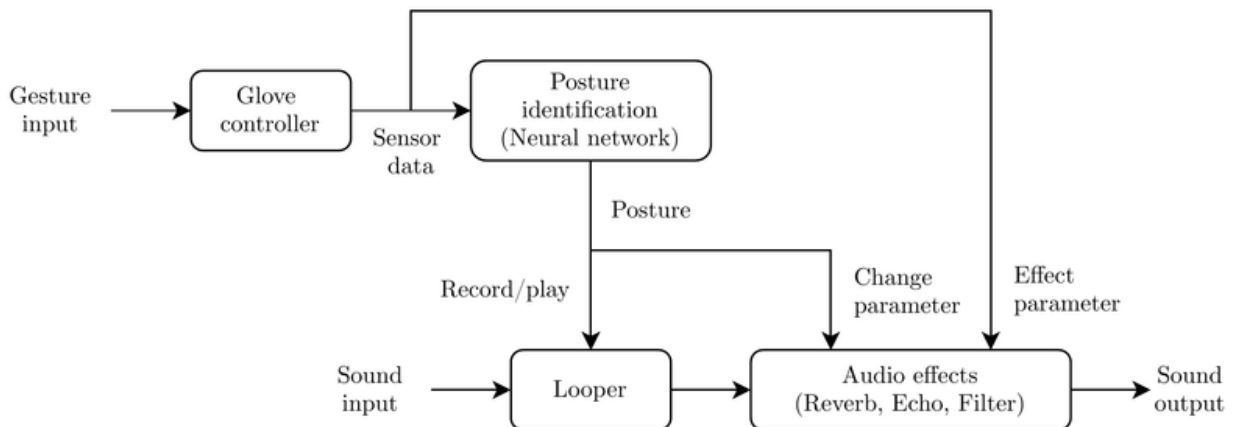

Image 2

Gesture-to-sound interface of *SoundGrasp*. Gestures are recognized using a neural network. The identified postures are used as commands for controlling the looper. Sensor data is also mapped directly to audio effect parameters.

Control stream loopers

Streams of control data such as MIDI or Open Sound Control (OSC) messages or analog control voltages (CV) can also be looped, by inserting the looping device between a control interface and a sound generator like a mapping layer. As with audio stream loopers, control data is recorded into a buffer and played back in a loop. Control stream loopers offer the advantage that mappings can be changed post-recording, giving the possibility to re-route the control data to different synthesis processes.

MidiREX and Midilooper

MidiREX [12] by Peter Kvitek and *Midilooper* [13] by Bastl Instruments (Image 3) take their inspiration from digital loop pedals both in appearance and functionality. The devices record incoming MIDI messages into a buffer, also compatible with MIDI Polyphonic Expression (MPE) [14]. Midilooper can modulate MIDI velocity either randomly or using a control voltage input as a modulation source. Random modulation has become an increasingly popular feature of music sequencers as a tool for "humanization" [15]—a trend Cascone characterizes as an era of "post-digital" music defined by the aesthetics of failure and audible glitches [16]. *Midilooper*'s random velocity feature, labeled "human velocity", can add dynamic variation to the recorded loops.
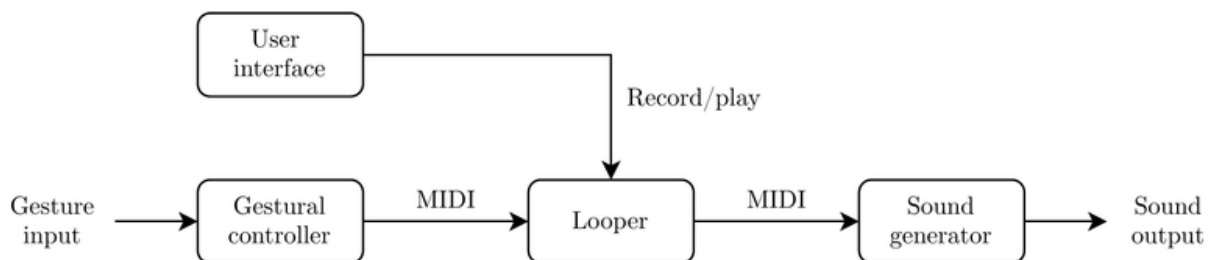

Image 3
Gesture-to-sound interface of *MidiREX* and *Midilooper*. The MIDI protocol allows using any MPE-compatible gestural controller.

Ribn and Tetrapad

Ribn [17] by Nobjsa Petrovic and *Tetrapad* [18] by Intellijel (Image 4) have touch interfaces to record horizontal or vertical gestures. Up to eight sliders can be added to Ribn's interface, with each sending a single MIDI control change message. Recording starts when the slider is touched and ends on release. Playback starts immediately after recording, and loop lengths can not be changed after recording. *Tetrapad* is a

Eurorack module with four touch zones that sense both position and pressure, allowing for two-dimensional gesture recordings. *Tetrapad* has eight control voltage outputs that can be patched to any parameter within a Eurorack system. With the *Tête* expander module, recorded sequences can be quantized in both time and value, with the possibility of quantizing control voltage outputs to a selection of musical scales.
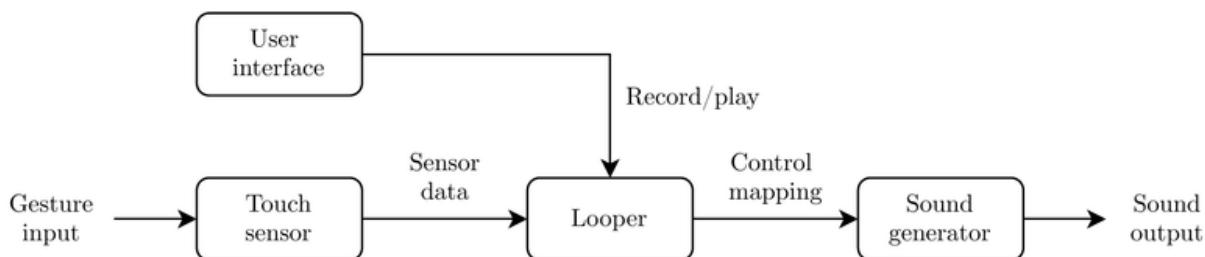

Image 4

Gesture-to-sound interface of Ribn and Tetrapad + Tête. Touch sensor is embedded in the interface.

Drile

Drile [19] by Berthaut et al. (Image 5) is a virtual reality-based live-looping system. A bi-manual 6-DoF controller is used to create loops and control audio effects in a 3D space. Unlike the other looping tools, *Drile* supports both audio and control streams, and offers hierarchical live-looping by grouping loop layers in a hierarchical tree instead of a flat structure. Loops can be layered per instrument or section in a piece.
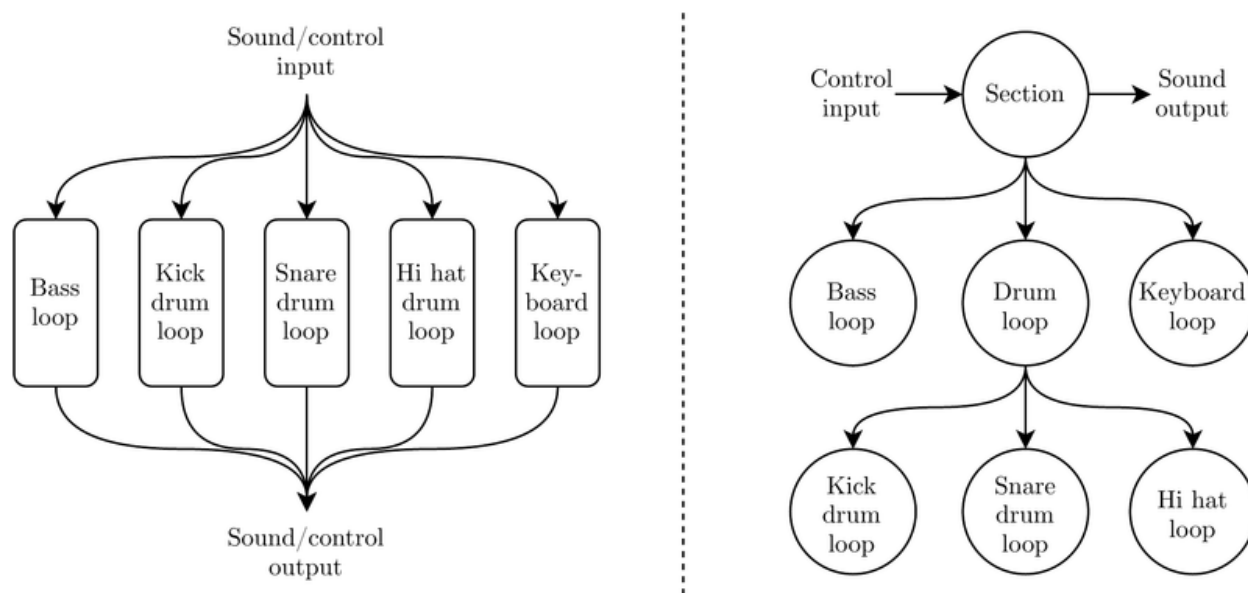

Image 5

Traditional and hierarchical live-looping structures with *Drile*.

Summary

We provide a comparison of related work versus our tool *MapLooper* in Table 1. *Interact* refers to the gestural interface where **(x)* means all devices supported by *x*. *Loop* refers to the interface for switching recording and playback state. *Quantize* refers to time quantization. *Manipulate* refers to any real-time processing of the recorded loops.

Table 1							
Project	Stream	Interact	Loop	Quantize	Synchronize	Manipulate	Map
SoundCatcher	Audio	Ultra-sonic	Footswitch	No	Yes	Audio FX	Explicit
SoundGrasp	Audio	Glove	Posture	No	No	Audio FX	Machine learning
MidiRex	Control	*(MPE)	Button	Yes	Yes	No	Explicit
Midilooper	Control	*(MPE)	Button	Yes	Yes	Random/CV	Explicit
Ribn	Control	Touch	Touch	No	No	No	Explicit
Tetrapad	Control	Touch	Touch	Yes	Yes	CV	Explicit
Drile	Both	6-DoF	6-DoF	Yes	No	No	Explicit
MapLooper	Control	* (*libmapper*)	* (*libmapper*)	Yes	Yes	Random	Open-ended

While most of the tools reviewed contain their own gestural interface; only *MidiRex* and *Midilooper* can use external gestural interfaces. However, with these two tools, the recording and playback state can only be controlled using a button. All of the reviewed tools feature either time quantization, external synchronization, or loop manipulation. Most of the tools' mapping strategies are explicit, except for *SoundGrasp*, which employs mapping using machine learning.

Design requirements

Our review guided the design requirements of our tool that should support:

- changing sound sources after recording,
- looping streaming data from different gestural controllers,
- controlling loops with open-ended gestural interfaces,
- quantizing time,
- synchronizing time externally,
- manipulating loops by random modulation,
- mapping with both explicit and machine learning strategies,
- running on a wireless embedded device,
- replicating its open-source components.

Infrastructure for embedded, distributed and synchronized mapping

To build applications for live-looping satisfying the design requirements that we elicited, we developed a connectivity infrastructure for wireless mapping and synchronization. We ported existing libraries for mapping and synchronization to a wireless embedded platform.

Embedded platform

For the wireless embedded platform, we use the ESP32 microcontroller: a small, cheap, and sufficiently powerful chip for digital signal processing [20].

Mapping framework

To build a looper with advanced mapping capabilities, we use the mapping software *libmapper* [21] as the main building block.

Embedded Mapping Components

We adapted *libmapper* and its dependencies to run on ESP32 platforms: we implemented functions in *compat-idf* for compatibility between pthread and the Free Real-Time Operating System (FreeRTOS), we ported the *liblo* library for OSC communication, and we compiled the *zlib* for data compression.
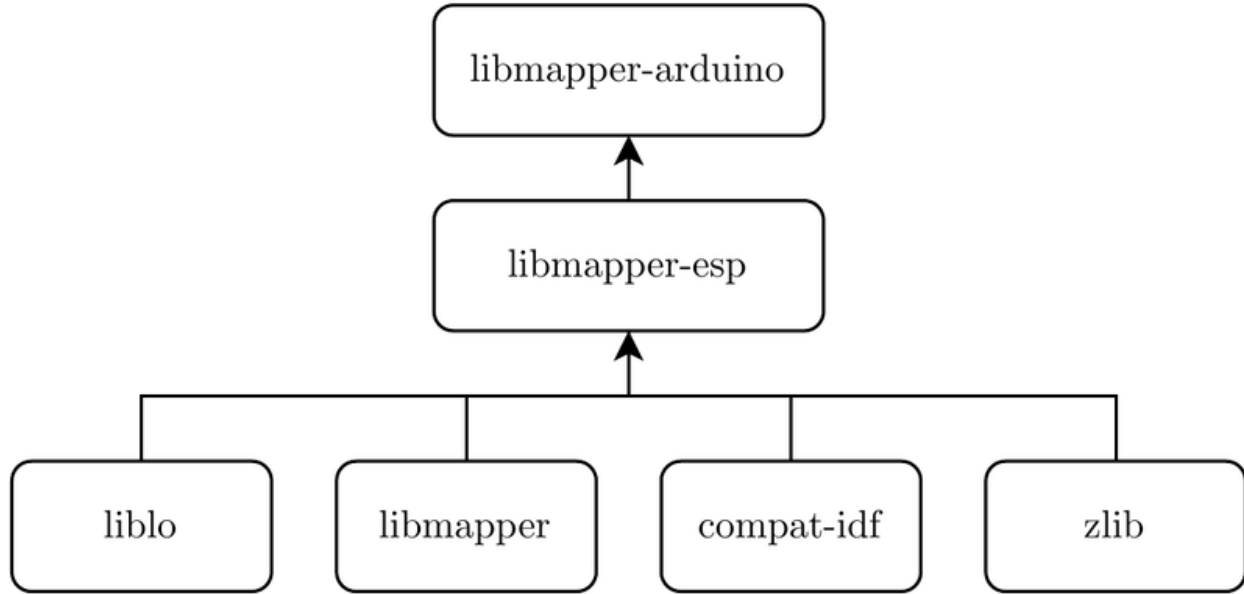
Image 6
Structure of the libraries ported to ESP32 for libmapper support.

liblo

The *liblo* library relies on POSIX sockets and threads (*pthreads*) for creating UDP/TCP sockets and servers. The Espressif IoT Development Framework (ESP-IDF) [22]) contains a pthread library that partially translates the FreeRTOS API into the POSIX threads API that we needed to update.

compat-idf

We implemented several POSIX functions that were missing for networking embedded DMIs (*getnameinfo*, *gai_strerror*, *gethostname*, *getifaddrs*, *freeifaddrs*) and packaged as an open-source ESP-IDF component, *compat-idf* [23].

libmapper-esp

We packaged these four components, *liblo*, *libmapper*, *compat-idf*, and *zlib*, as an open-source ESP-IDF component, *libmapper-esp* [24].

libmapper-arduino

To facilitate embedding *libmapper* support in DMIs like the T-Stick DMI using common Integrated Development Environments, we implemented an Arduino version of the *libmapper* library that we release as the open-source *libmapper-arduino* library [25].

Testing

We measured round-trip latency, jitter, and package loss for data transmitted through embedded *libmapper*. Our test setup consisted of applications running on two computing devices. 1) The firmware of an ESP32 WROVER KIT development board [26] running the *libmapper-esp* library creates one input and one output signals. The input signal handler is set to forward incoming data to the output signal. 2) A software application running on a MacBook Pro laptop (16-inch, 2019, macOS 10.15) sends a 100 Hz signal to the ESP32, and we measure the time between sending and receiving data. The ESP32 was running in access-point mode and the computer was connected to this access-point through WiFi. The results are in Image 7 .
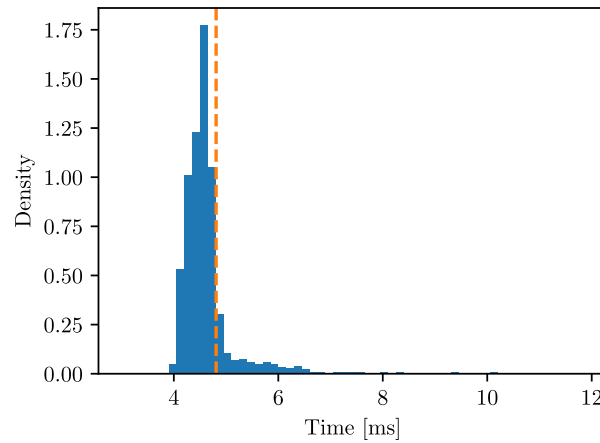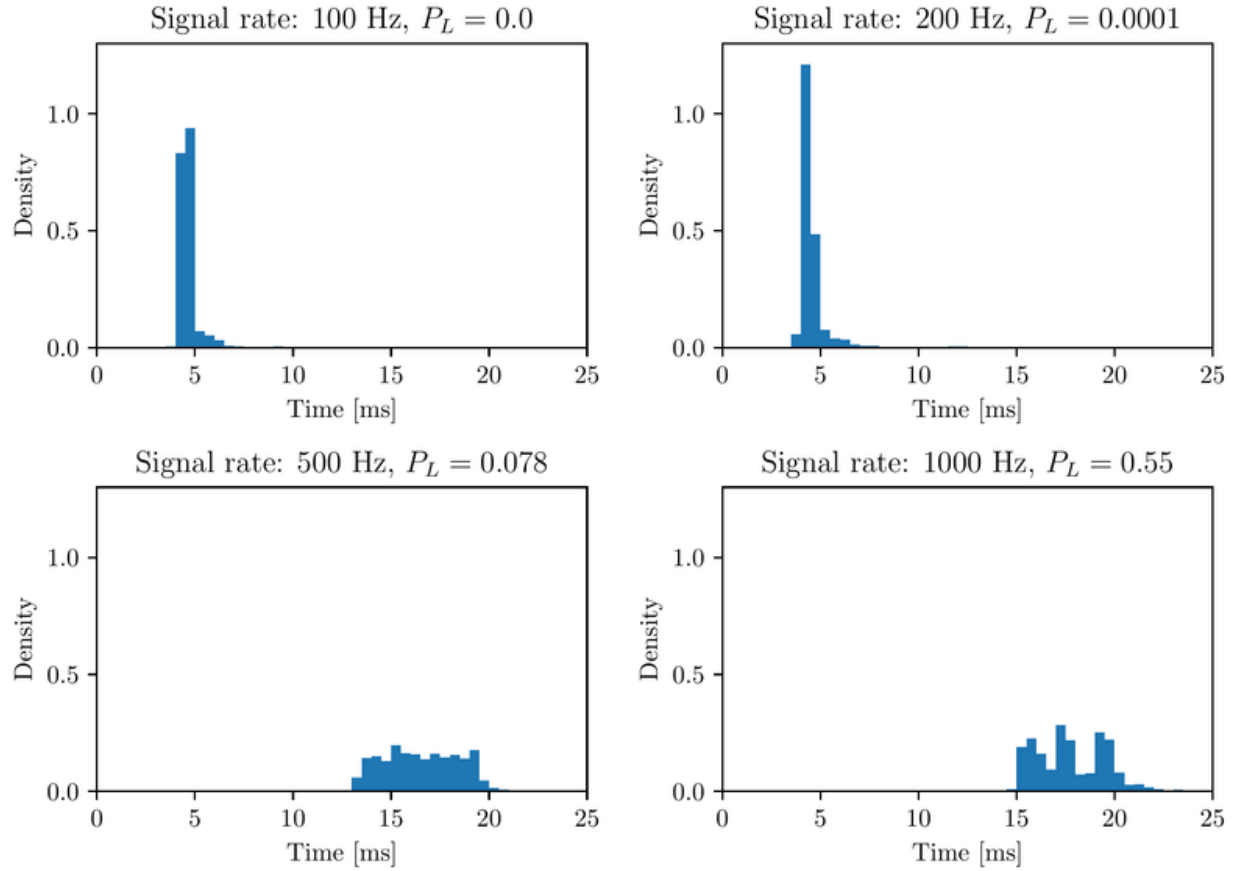

Image 7
Histogram of round-trip latency
measurement with power saving feature
disabled. A 100 Hz is signal measured
over a period of 100 seconds. The
dashed line shows the mean round-trip
latency of 4.81 ms.

We found that the ESP32 has a WiFi power-saving feature enabled by default. Disabling this feature had a significant impact on low latency performance. We performed measurements with power-saving enabled or disabled. The mean of the round-trip latency was 406 ms with power saving enabled and 4.81 ms when disabled. According to our results, in a one-way communication situation, where the ESP32 is only transmitting data, an average end-to-end latency $L_m = 2.41$ ms (half of round-trip) can be expected.

We performed three more measurements at increasing rates for testing latency, jitter; and packet loss P_L at different signal rates. The results are in Image 8.

Image 8

Histograms of round-trip latency of test signals at 100 Hz, 200 Hz, 500 Hz, and 1000 Hz, with 10,000 test points recorded for each signal. Packet loss P_L and frequency is listed in the title of each histogram.

We found that the system has a significant packet loss for signals at 500 Hz. For signals at 1000 Hz, the packet loss is substantial, with 55% of packets being dropped. The jitter also increases with frequency, which can be observed in the increase of the standard deviation of the latency listed in . There is no significant change in the mean latency for signals at 100 Hz and 200 Hz. For signals at 500 Hz, the latency increases by a factor of 3. Signals at 500 Hz and 1000 Hz had similar performance in terms of latency and jitter, but the packet loss increases from 7.8% at 500 Hz to 55% at 1000 Hz. Table 2 provides results from latency measurements.

Table 2			
Signal rate [Hz]	Mean latency [ms]	Std. dev. of latency [ms]	Packet loss P_L

100	4.81	1.56	0.0
200	4.78	1.86	0.0001
500	16.6	1.92	0.078
1000	17.9	1.98	0.55

Our results for the embedded *libmapper* implementation were slightly better than previous studies by Wang et al. [27], who conducted tests of latency and jitter with OSC communication over WiFi using ESP32. They measured a mean round-trip latency of 6.62 ms, which is slightly higher than the 4.81 ms we measured in this project, both at 100 Hz. Both measurements remain well below the "acceptable upper bound on the computer's audible reaction to gesture at 10 ms" proposed by Wessel and Wright [9].

Synchronization framework

For time synchronization between devices on a wireless network, we ported Ableton Link [28]: an open-source library for synchronizing tempo, beat, phase, and start/stop commands. Turchet et al. mention Ableton Link as a candidate for becoming a standard for music synchronization for Internet of Musical Things (IoMusT) devices [29].

Embedded Synchronization Components

To compile and run Ableton Link on ESP32, we needed to port the following modules to FreeRTOS:

- Clock : a simple timer with microseconds resolution.
- Context for the asynchronous operation of Ableton Link.
- LockFreeCallbackDispatcher for real-time safety of the session state.
- Random for random identification string generation for the peer.
- ScanIpIfAddrs for retrieving information about the available network interfaces on the system.

We distribute this library as an open-source ESP-IDF component: link-esp [30].

Testing

To test our embedded port of Ableton Link, we created a test setup for measuring the delay between peers. The test setup consisted of two MacBook Pro laptop computers

(Computer 1: 16-inch, 2019 and Computer 2: 15-inch, 2018; both running macOS 10.15) and an ESP32 board, all connected to a RIGOL DS1054 oscilloscope. Two probes were connected to an audio jack from the headphone output of each of the computers. One probe was connected to a GPIO pin of the ESP32. The computers synthesized a pulse signal through Ableton Live [31]. The ESP32 ran a test software outputting a pulse on a GPIO pin. All devices were connected through an Ableton Link session and outputted a periodic pulse on every quarter note at 120 BPM. A plot of the measurements is in Image 9.

We found that the ESP32 performs similarly to the two laptop computers in terms of inter-onset delay. Over 10 minutes, the average delay between Computer 1 and ESP32 was 3.03 ms (min: -6.62 ms, max: 0.02 ms).
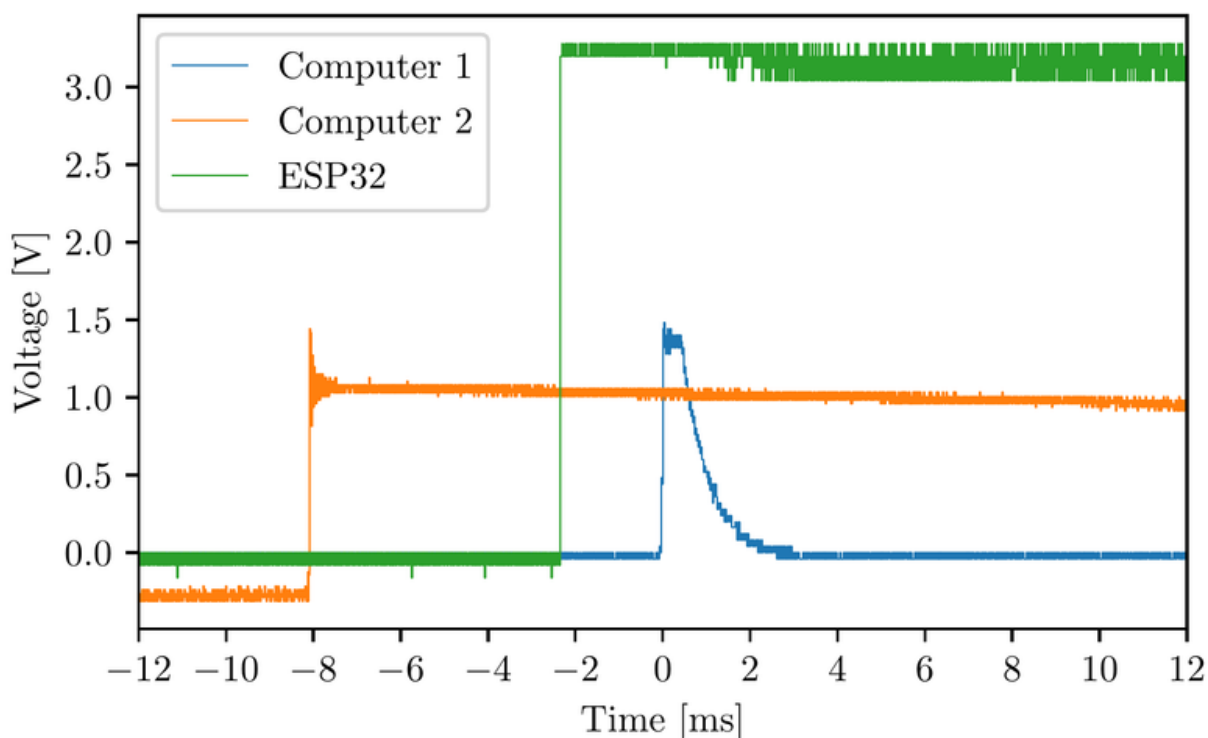

Image 9
Oscilloscope measurement of an Ableton Link session consisting of two computers and an ESP32. All peers output a pulse signal at every quarter note at 120 BPM.

Application: implementation of *MapLooper*: gesture-to-sound looper

This section describes *MapLooper*, our gesture-to-sound looping tool built upon our connectivity infrastructure. We implemented *MapLooper* based on a delay-line model

using *libmapper* map expressions. We present two musical applications built with the tool. We distribute *MapLooper* as an open-source software [32].

Looping with a delay-line

We can build a digital looper by adding feedback to a delay-line. A digital delay-line is a special case of IIR filtering, which is supported by *libmapper* for exponential smoothing. The discrete-time system implementing a digital looper can be expressed in terms of a linear interpolation between an input $x[n]$, and a delayed output term $y[n - D]$, with the linear interpolation factor representing a record signal $r[n]$ so that: $y[n] = r[n] \cdot x[n] + (1 - r[n]) \cdot y[n - D]$. A block diagram of this system is in [Image 10](). For most live-looping devices, the record/playback state is boolean, and the signal $r[t]$ is either 0 or 1. When $r[t] = 0$, only the delay-line output is passed to the system's output. When $r[t] = 1$, the input is passed directly to the output and into the delay-line, thereby being recorded. For $0 < r[t] < 1$, overdub can be achieved as the input is mixed with the delayed input.
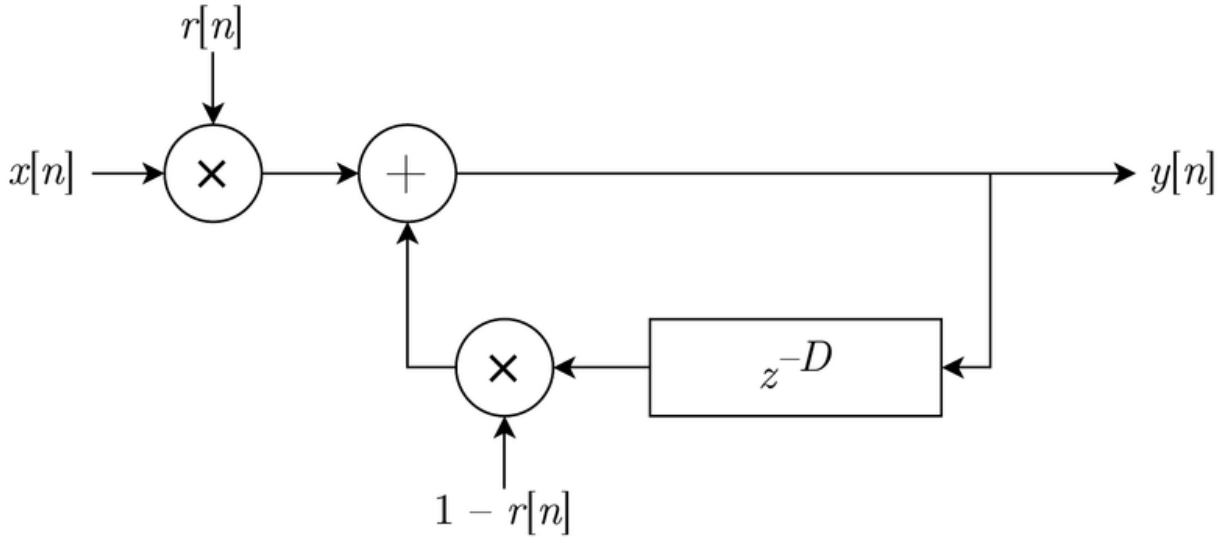

Image 10
Block diagram of basic looping system implemented as a delay-line with feedback.

Synchronization and time quantization

For a loop to be synchronized to a meter, the length D of the delay-line should be specified in terms of tempo [bpm] T and duration in beats B

$$D = \frac{B \cdot 60}{T} \tag{1}$$

For a 1-bar loop with a tempo of 140 bpm and time signature 4/4, this results in

$$D = \frac{4 \text{ beats} \cdot 60 \text{ s}}{140 \text{ beats/min}} = 1.7142857143 \text{ s} \tag{2}$$

At the time of our initial implementation, delay-lines in *libmapper* were non-interpolating in terms of delay-length. We first solved this issue by sampling the input at a rate given by an integer subdivision of the tempo, ensuring that delay-lengths were always an integer multiple of the loop-length in beats. We have since added fractional delay lengths to *libmapper*.

We added a sample-and-hold structure to the system to implement tempo-synchronized sampling. A clock signal $c[t]$ synchronized with the tempo triggers a sampling of the input signal $x[t]$. The rate of the clock determines the quantization. This rate is commonly given for analog synchronization systems in the unit *pulses per quarter note* (PPQN). A block diagram of this system is in Image 11.
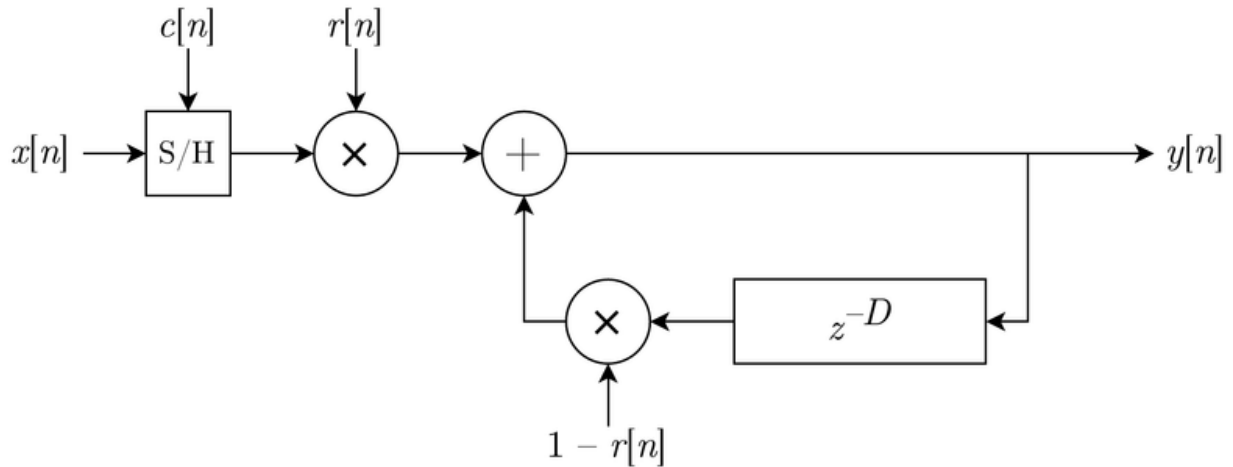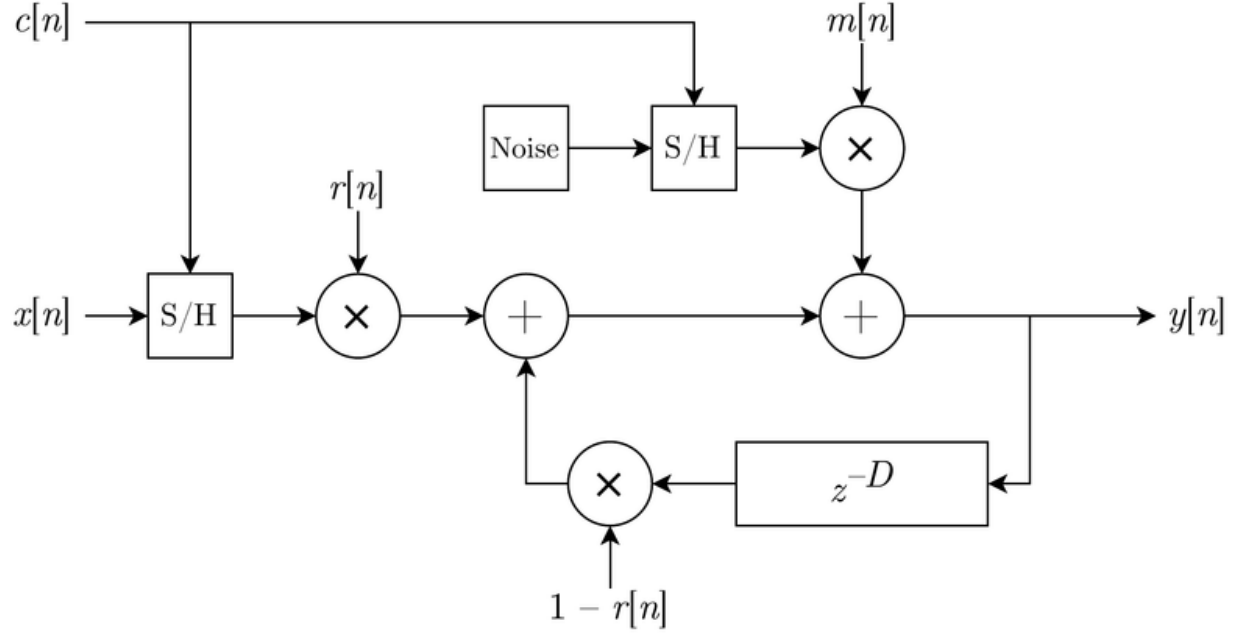

Image 11
Block diagram of looping system with time quantization.

Loop manipulation

We implemented a simple modulation system modelled on the sample-and-hold structure. We used a uniform noise generator as a modulation source, sampled at the same rate as the input. We added this modulation signal within the feedback path, so that an input sequence could be recorded, after which modulation could be applied to make the sequence slowly evolve over time. A block diagram of the system is in Image 12.

Image 12
Block diagram of loop manipulation system. The loop is modulated by noise through a sample-and-hold structure. The modulation is within the feedback path.

The uniform noise generator creates a noise signal with a range between [-1, 1] multiplied by the signal $m[t]$, controlling the modulation amount. For small amounts of modulation, the original contour of a recorded sequence is retained on a macro timing level with an increasing variation on the micro timing level.

Control signals

MapLooper instantiates a *libmapper* device and initiates a Link session. The control interface consists of five signals ([Table 3](#)): *record*, *length*, *division*, *modulation*, and *mute*.

Table 3				
Signal	Description	Unit	Min	Max
record	Controls whether input is active	-	0	1
length	Length of the loop	beats	1	100*
division	Time quantization	PPQN	1	100

modulation	Amount of modulation	-	0	1
mute	Controls whether output is active	-	0	1

The *record* signal represents the $r[t]$ signal in . The *length* and *division* signals determine the length D of the delay-line by the relation: $D = \text{length} \cdot \text{division}$. The *length* signal is limited by the current maximum of 100 samples of delay in *libmapper*, though the library can be recompiled with additional memory. The *modulation* signal represents the $m[t]$ signal in . The *mute* signal was added to control whether the output from *local/recv* propagates to the *output* signal.

For each loop instance, a convergent map is created between the control signals, the *local/send*, and the *local/recv* signal. A map expression is created for the map, describing the system in Image 13.
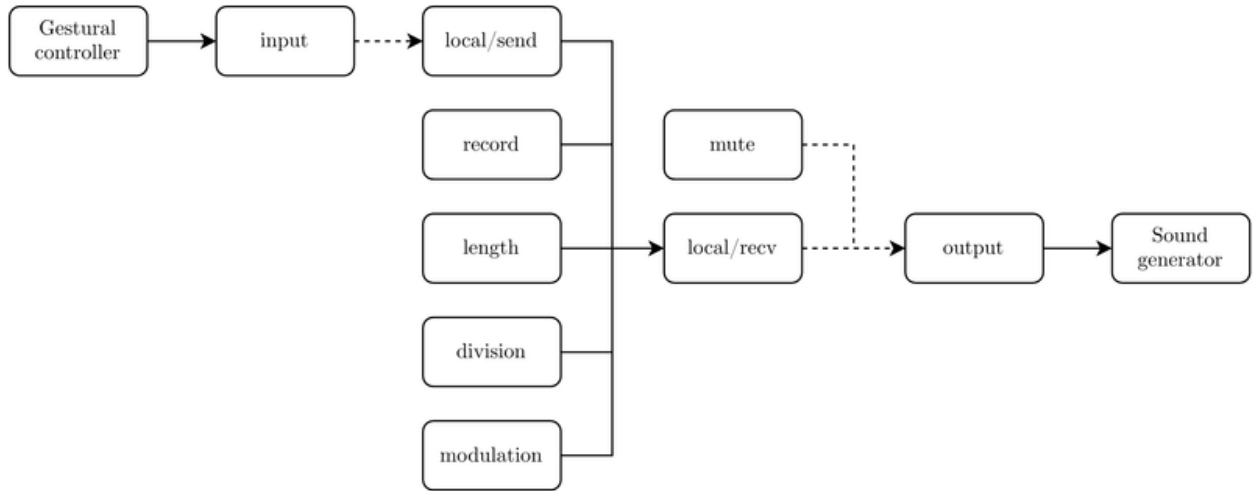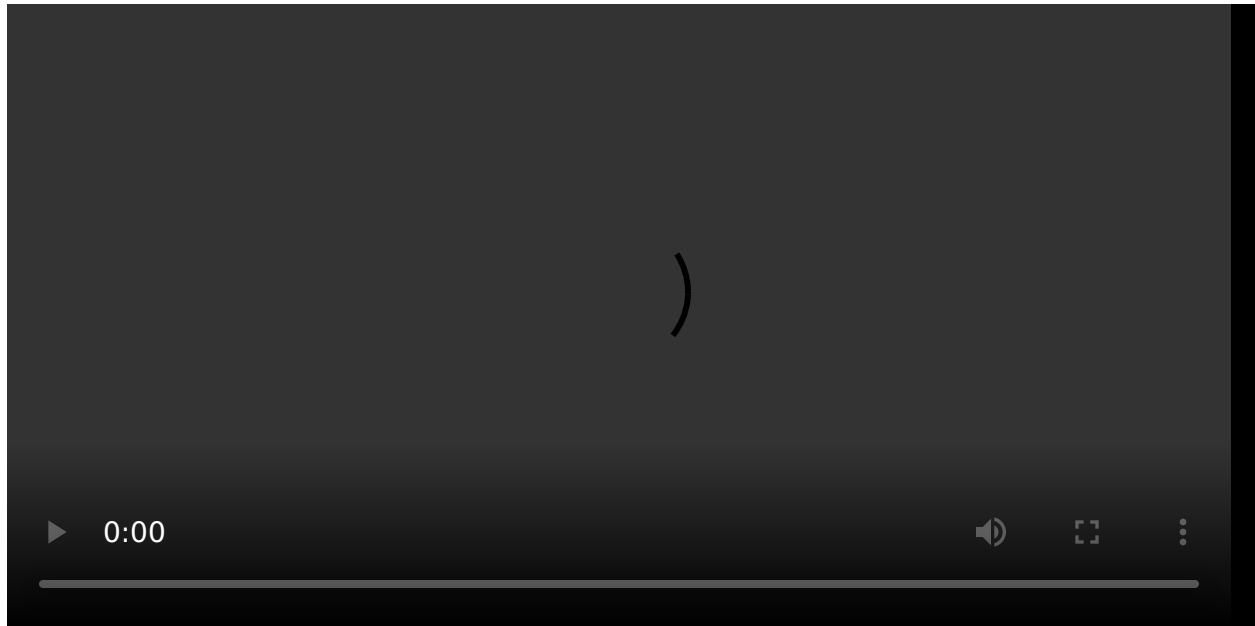

Image 13

Block diagram of mapping configuration. Each block represents a signal. The solid lines represent the convergent mapping between the control signals and the local send and receive signals. The dashed lines are internal mappings done outside of *libmapper*.

During loop updates, the input is sampled at a rate synchronized with Link. The sampled value is sent to the *local/send* signal, and the map expression is evaluated. Finally, if the Loop instance is not muted, the value of the *local/recv* signal is copied to the output signal.

By mapping a gestural controller to the input signal and a sound generator to the output signal (Video 1 and Image 14), a DMI with looping capabilities can be implemented.

Video 1
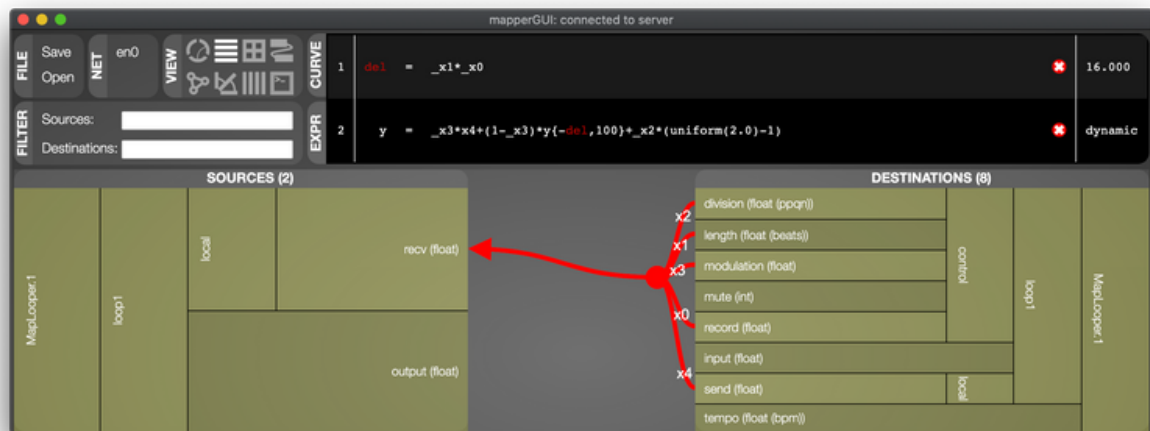Construction of the mapping in visualization tool Webmapper.

Image 14
Visualization of the mapping in visualization tool Webmapper.

Graphical User Interface

For testing, we created a cross-platform GUI application based on the JUCE framework [33], containing six sliders and a button (Image 15). When launching the GUI, a loop instance is created, and sliders are initialized to the default values of control signals. Slider 1 *input* sends its value to the loop's input, and similarly to Ribn and Tetrapad the value of the slider is only recorded when the slider is pressed. Slider 2 *output* displays the output of the loop and is not editable. The remaining four sliders control: the loop length in beats (slider 3), the amount of noise modulation (slider 4), division in pulses per quarter note (slider 5), and tempo in beats per minute (slider 6). A toggle at the bottom controls whether the local loop map's output propagates to the loop's output. We distribute *MapLooper-gui* as an open-source project [34].
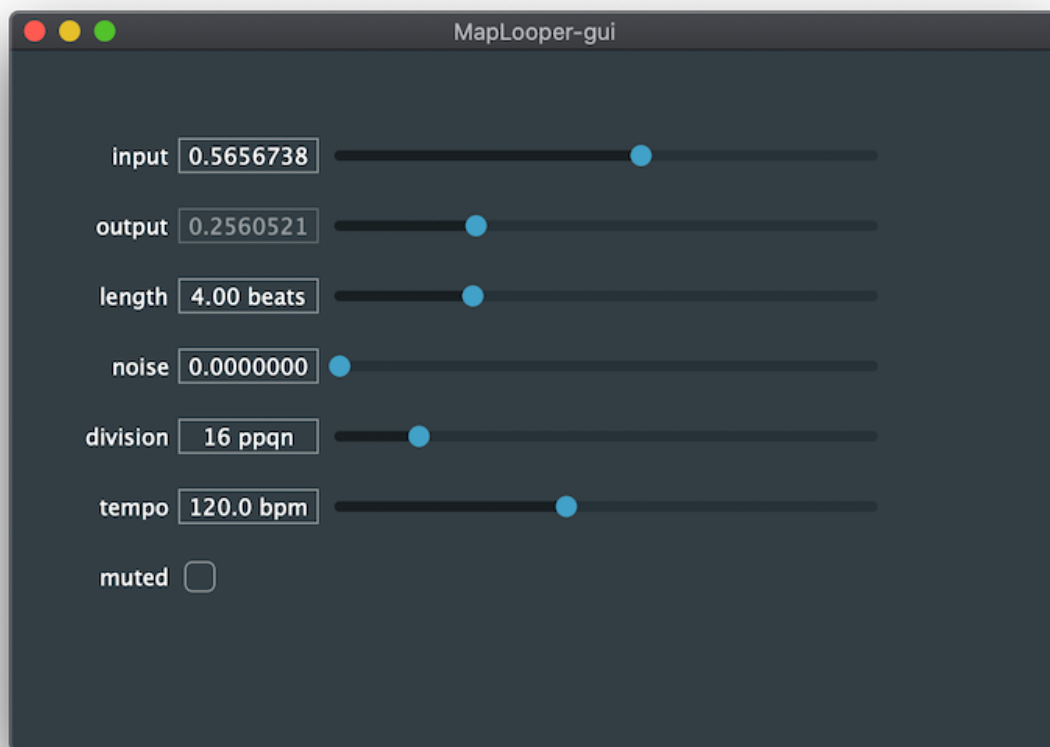

Image 15
Screenshot of JUCE-based *MapLooper-GUI*.

Sound synthesis examples

SuperCollider extension: *MapperUGen*

We implemented a new SuperCollider UGen server extension called *MapperUGen* [35] for using *libmapper*. The extension has classes for creating input and output signals (MapIn and MapOut) with signal names and ranges specified as arguments for the constructor. When synths are created and destroyed in SuperCollider, UGens are erased from memory, which causes maps to SuperCollider to be destroyed. We implemented a system for persistent maps by saving *libmapper* signals in a global variable. When MapIn and MapOut are instantiated, the classes automatically bind to existing signals with the signal name given as an argument. This solution optimizes the workflow considerably when prototyping mappings.

Harp demo

We created one musical demo by mapping the output signal to a harp synthesizer implemented in SuperCollider ([Video 2](#)).
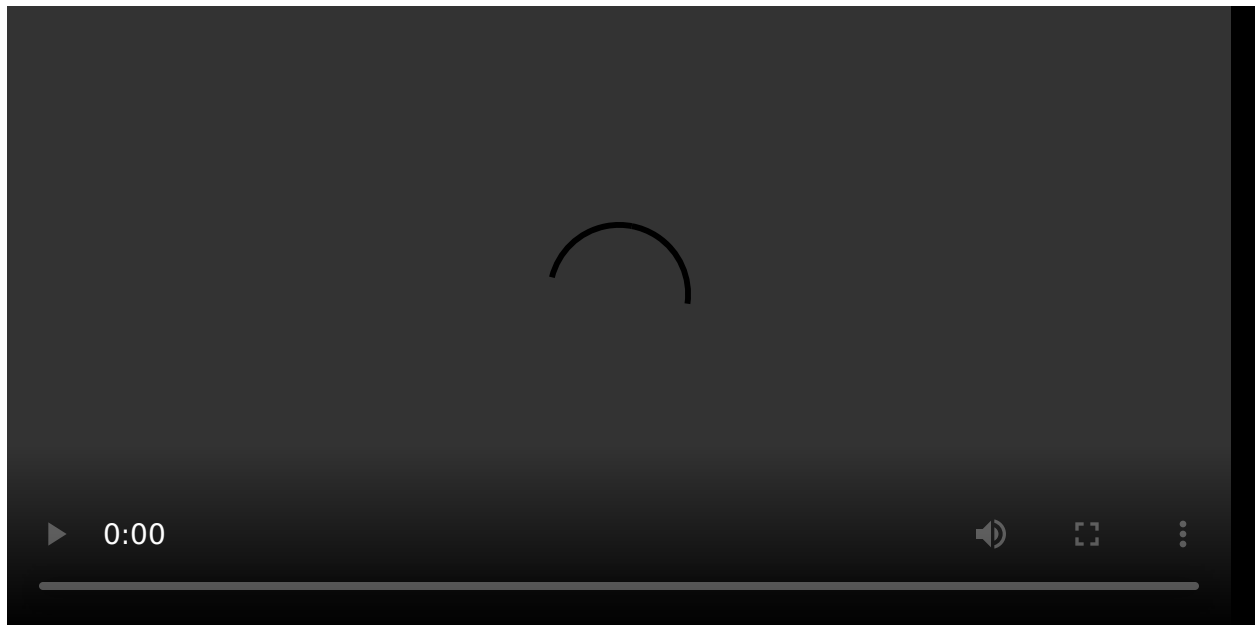
Video 2
Demo with harp synthesizer implemented in SuperCollider.

The harp synthesizer is based on a Karplus-Strong string model. Two input signals, *frequency* and *amplitude* control the frequency and amplitude of the string model. The frequency input is quantized to a melodic scale within the synthesizer, and a slope detector on the quantized frequency triggers the string excitation. As a result, when

moving the input slider, melodic notes are triggered along with the range of the slider. The interaction provides a similar feel as when sliding fingers over the strings of a harp. The SuperCollider code for the demo is:

```
fork {
  Mapper.enable;
  // Wait 2 seconds for libmapper initialization
  2.wait;
  {
    var index, scale, freqCtl, freq, amp, src, trig;
    // Create buffer with pentatonic minor scale
    scale = 36.collect{ |i|
      Scale.minorPentatonic.degreeToFreq(i, 50, 0);
    }.as(LocalBuf);
    // libmapper input signals
    freq = MapIn.kr(name: \freq, min: 50, max: 2000);
    amp = MapIn.kr(name: \amp, min: 0, max: 1);
    // Quantize frequency to pitch
    freq = Index.kr(bufnum: scale, in: IndexInBetween.kr(scale, freq));
    // Trigger the string on change
    trig = Changed.kr(freq);
    // Karplus-Strong string model
    src = Pluck.ar(in: PinkNoise.ar, trig: K2A.ar(trig), delaytime: 1 / freq);
    src * 0.5;
  }.play;
}
```

Embedded Sound Synthesis

We also created a proof-of-concept demo of using the looper with embedded sound synthesis (Video 3).
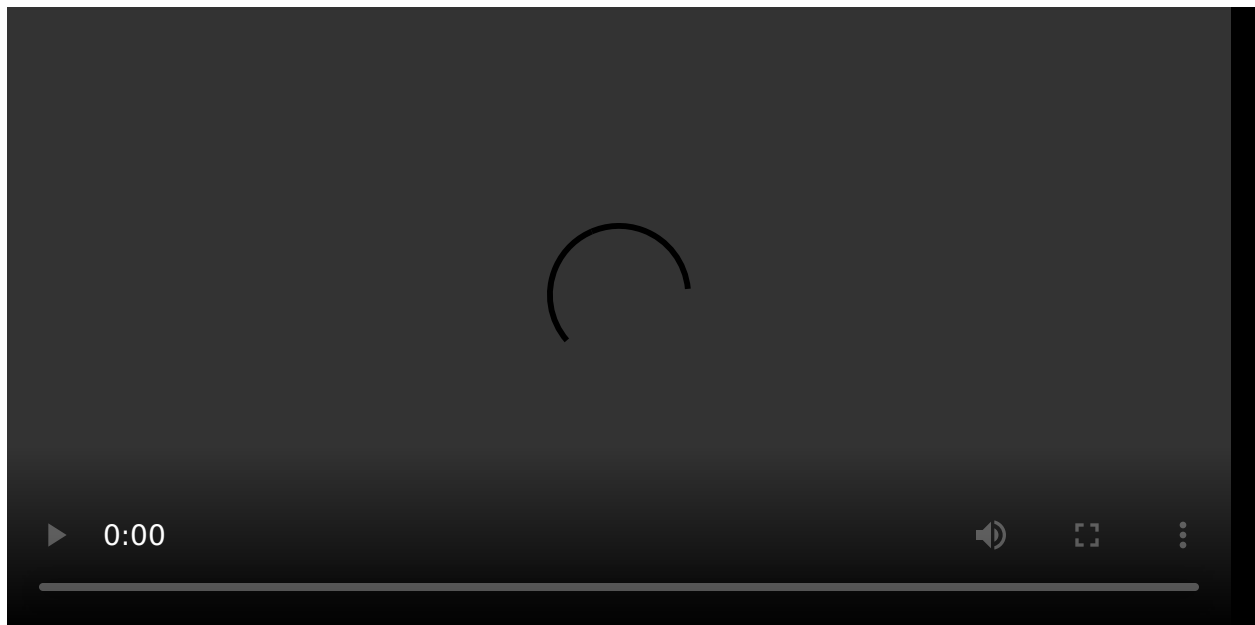
Video 3
Embedded demo with pink noise passed through a Moog-style voltage-controlled filter emulation.

The demo was based on the ESP32 LyraT board [36] (Image 16), which contains an ESP32 WROVER module and an audio codec chip along with 1/8 inch TRS connectors for headphones and auxiliary audio input.
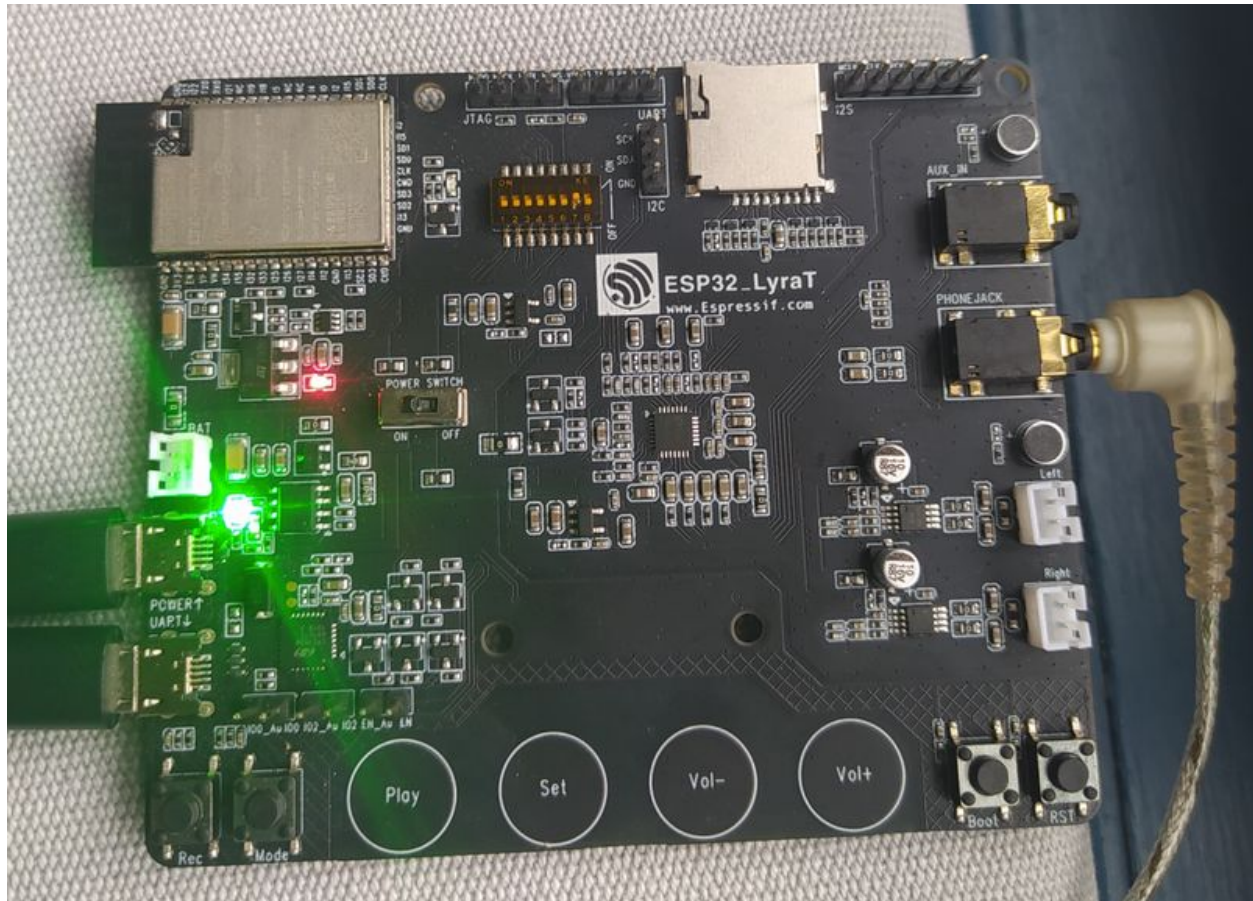

Image 16
The LyraT board.

We release our demo [37] as an open-source project using the Faust library [38] for compiling a DSP program to the LyraT board, which is supported by the Faust compiler [39]. The DSP program used for the demo is:

```
import("stdfaust.lib");
ctFreq = hslider("cutoffFrequency", 500, 50, 3000, 0.01);
res = hslider("resonance", 0.5, 0, 1, 0.1);
gain = hslider("gain", 1, 0, 1, 0.01);
process = no.pink_noise : ve.moog_vcf(res, ctFreq) * gain;
```

The program generates pink noise and passes it through a Moog-style voltage-controlled filter emulation. The program has three parameters: cutoff frequency, filter resonance, and output gain. A Loop is created for each parameter mapped to a

libmapper signal that updates the parameter when receiving a value. A random number generator sends an input signal to each of the loop layers. The recorded signal is 1.0 when the program starts and is set to 0.0 after 10 seconds. The program continues indefinitely, repeating the same 1 bar sequence. A block diagram of the demo program is in Image 17.
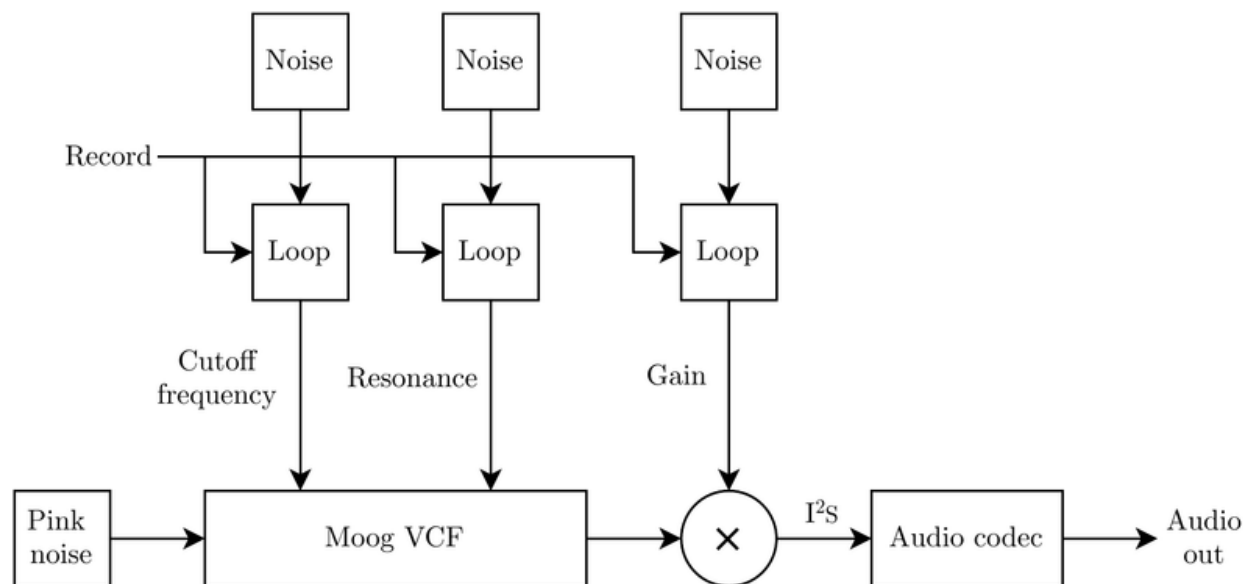

Image 17
Block diagram of embedded sound synthesis example.

Conclusion and future work

We have presented the development of a live-looping system for gesture-to-sound mappings built on a connectivity infrastructure for wireless embedded musical instruments with distributed mapping and synchronization. We evaluated in the context of the real-time constraints of music performance: round-trip latency, jitter, and package loss of signals transmitted through embedded mapping; inter-onset delay between peers for networked looping synchronization. On top of this infrastructure, we developed *MapLooper*: a live-looping tool with example musical applications: a harp synthesizer with SuperCollider and embedded source-filter synthesis with FAUST on ESP32.

We follow by discussing perspectives on our work.

Scalability and flexibility of map expressions

Implementing our system using *libmapper* brings scalability, support for vector signals, signal instances [40] and freely mixing mapping and looping. The map expression

interface allows for flexible mapping configurations for loop manipulation. The random modulation implementation could be changed to use any modulation signal by merely changing the map expression.

Delay-line models: continuous signals, zipper noise

One limitation is that the delay-line based model only allows for continuous signals. The delay-line model updates the output at every time quantization step. Continually updating the output can be an issue in scenarios where event-based signal updates are needed.

Additionally, the delay-line model causes issues when changing loop-length known from echo effects as zipper-noise. This noise is caused by discontinuities in the signal when adjusting the read pointer of a circular buffer. For interpolating delay-lines, the zipper-noise is replaced by Doppler-shifts. This effect has been used creatively as an audio effect, but it might not be what users expect for control data streams. The issue could be solved by cross-fading multiple read pointers when the loop-length is changed.

Latency compensation

When gestures are recorded through *libmapper*, all samples are time-tagged. Latency could be subtracted during playback to achieve accurate timing. Peers could continuously measure the latency between them by periodically sending a heartbeat signal and keeping a record of each peer's round-trip latency. This idea is similar to how host time offsets are handled with Ableton Link. At sampling frequencies above 500 Hz, our implementation had significant reliability issues. Instead of networking all peers at a high sampling rate, each peer could locally acquire the gestural data at a higher sampling rate while only sending quantized data to the network.

Visual and haptic feedback

When recording gesture-to-sound sequences in our looper, the instantaneous feedback gets lost once the recording is finished, since the auditory feedback no longer corresponds to the physical gesture currently being held. In the case of a single loop layer recording, our *MapLooper-GUI* provides visual feedback through a slider that displays the current output value of a loop. However, for more complex mappings, where more layers are being recorded simultaneously, the current system provides no feedback on what has been recorded. This missing feedback could be in the form of visualization on a screen, displaying multiple recorded sequences simultaneously. The loop visualization tool could be developed as an extension of WebMapper. Additionally,

feedback could be given in the form of haptic feedback, for instance with *TorqueTuner*[41] also embedding *libmapper*, to display force cues mapped to recorded sequence.

Multiple read pointers

Multiple variable-speed read pointers could be implemented to explore new looping techniques inspired by multi-tap delays and granular time-stretching audio effects. A single loop layer could control several voices by mapping the instances to voices on a polyphonic synthesizer, adding variations on a micro time-scale. Non-constant time quantization could add the shuffle effect popular on many drum machines and featured in MidiLooper.

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