

An Open-Source C++ Framework for Multithreaded Realtime Multichannel Audio Applications

Matthias Geier¹ Torben Hohn² Sascha Spors¹

¹Quality and Usability Lab
Telekom Innovation Laboratories
Technische Universität Berlin

²Linutronix GmbH

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APF

Audio Processing Framework

- collection of C++ code
- <http://tu-berlin.de/?id=apf>
- open source, GPLv3
- mostly platform-independent
 - tested on Linux and MacOSX
- quite generic
- heavy use of templates
- mostly header-only

APF Components

- MimoProcessor → *topic of this talk!*
- biquad & cascade of biquads (IIR filter)
 - several methods for denormal prevention
- (yet another) C++ wrapper for JACK
- iterators
- other tools
- delay line → *coming soon!*
- partitioned convolution (FIR filter) → *coming soon!*

MimoProcessor

Target Applications

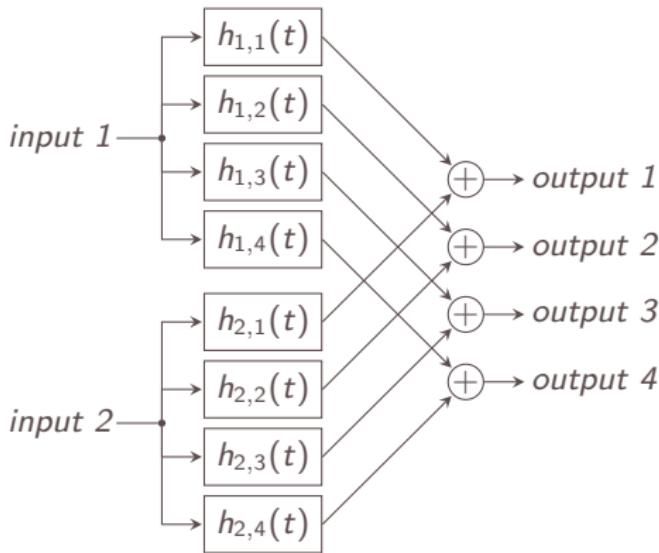
- block-based audio applications
- many inputs/outputs
- both realtime and non-realtime
- interactive applications

example application areas:

- sound field synthesis
- multichannel echo cancelling
- beamforming

MimoProcessor Example Application

Generic MIMO system



MimoProcessor Features

- different audio backends (realtime and non-realtime)
 - JACK
 - Pure Data (and Max/MSP) external using *flext*
 - MEX-file for octave (and Matlab)
 - read/write multichannel audio files
 - PortAudio → *coming soon!*
 - ... and users can provide their own backends!
- “automatic” multithreading
- optional crossfade
- dynamic number of inputs/outputs (if supported by the audio backend)
- safe communication between realtime and non-realtime threads

Realtime & Non-Realtime Threads

- realtime thread:

- audio callback function
- limited computation time per audio block
- blocking functions are **forbidden**, e.g.
 - allocating/deallocating memory
 - reading/writing files/sockets
 - creating/joining threads
 - waiting for mutexes
 - ...

- non-realtime thread:

- GUI, network, reading/writing files
- everything else

MimoProcessor Components

- LockFreeFifo<Command*>
 - push(), pop()
- CommandQueue
 - using 2× LockFreeFifo
 - push()/wait(), process_commands()
- Command
 - abstract base class
 - execute(), cleanup()
- RtList<Item*>
 - using CommandQueue
 - add()/rem(), begin()/end()/size()
- Item
 - abstract base class
 - process()

MimoProcessor Components

- MimoProcessor<Derived, *policies see below* >
 - interface_policy
 - jack_policy
 - pointer_policy
 - thread_policy
 - posix_thread_policy
 - sync_policy
 - posix_sync_policy
 - xfade_policy
 - raised_cosine_policy (default)
 - disable_xfade

Code Example

examples/jack_minimal.cpp

```
#include "apf/mimoprocessor.h"
#include "apf/jack_policy.h"
#include "apf posix_thread_policy.h"
#include "apf posix_sync_policy.h"

class MyProcessor : public apf::MimoProcessor<MyProcessor
    , apf::jack_policy, apf::posix_thread_policy, apf::posix_sync_policy>
{
public:
    typedef MimoProcessorDefaultInput Input;
    class Output;

    MyProcessor();

    void process()
    {
        _process_list(_output_list);
    }

private:
    rtlist_t _input_list, _output_list;
};
```

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```
class MyProcessor::Output : public MimoProcessorOutput
{
public:
    typedef MimoProcessorOutput::Params Params;

    explicit Output(const Params& p)
        : MimoProcessorOutput(p)
        , _combiner(_parent._input_list, _internal, _parent)
    {}

    virtual void process()
    {
        _combiner.copy(my_predicate());
    }

private:
    struct my_predicate
    {
        // trivial, all inputs are used
        bool operator()(const Input&) { return true; }
    };

    combine_channels<rtlist_t, Input, InternalOutput> _combiner;
};

};
```



```
MyProcessor::MyProcessor()
: MimoProcessorBase()
, _input_list(_fifo)
, _output_list(_fifo)
{
    Input::Params ip;
    ip.parent = this;
    _input_list.add(new Input(ip));
    _input_list.add(new Input(ip));

    Output::Params op;
    op.parent = this;
    _output_list.add(new Output(op));

    this->activate();
}

int main()
{
    MyProcessor processor;
    sleep(30);
}
```

Parallel Processing

- `RtList<Item*>`: list of polymorphic base class pointers
- virtual function `Item::process()`
- items within one list are processed in parallel
- fixed number of threads, specified by user
- simple scheduling:
 - each of the N threads gets every N -th item
- one “main audio thread”, $N - 1$ “worker threads”
 - communication via semaphores

Crossfade

- block-based processing
- parameter changes only at block boundaries
- artifacts due to discontinuities
- can be reduced by crossfade
- each block is processed twice
 - 1 with previous parameters, fade out
 - 2 with current parameters, fade in
- but: only if something actually changes
 - as noticed by CommandQueue
- crossfade is optional
 - can be switched off at compile time
 - `MimoProcessor<..., disable_xfade>`

Example

Near-Field-Compensated Higher Order Ambisonics

- implementation of a realtime NFC-HOA renderer
- circular loudspeaker array (2.5D)
- M -th order ($2 \times M + 1$ loudspeakers)
- stages of the algorithm (in RtLists):

N sources/inputs

$N \times (M + 1)$ IIR filters

$M + 1$ objects which add contributions of sources per order

- multiplication with 2 complex weighting factors
- resulting in $2 \times M + 1$ values (per audio sample)

block size IFFTs (of length $2 \times M + 1$)

$2 \times M + 1$ outputs

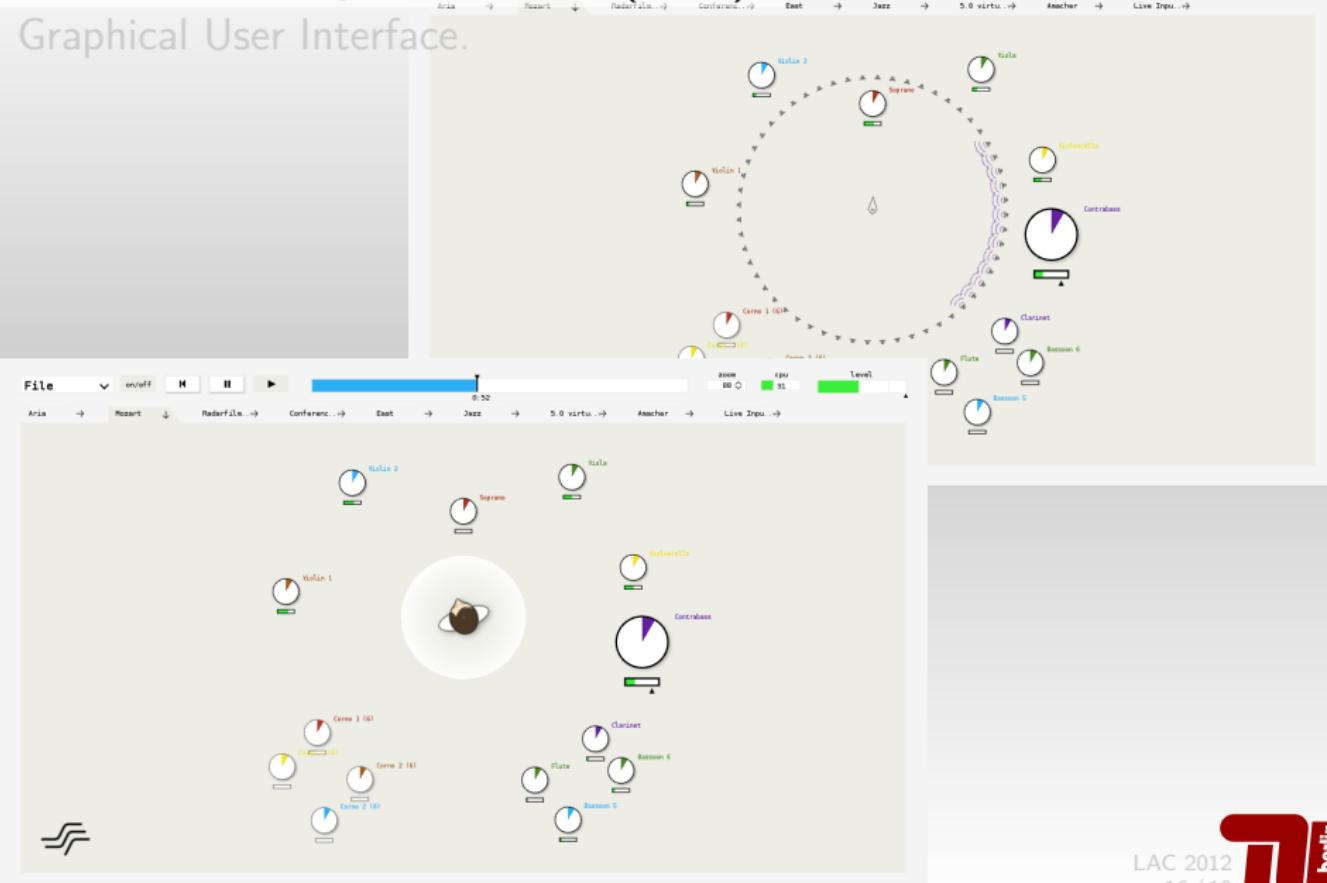
- part of the SSR → *coming soon!*
 - What is the SSR? → see next page

The SoundScape Renderer (SSR)

- software tool for *object-based* realtime spatial audio reproduction
- several different reproduction methods
 - Binaural Renderer
 - Binaural Room Synthesis (BRS)
 - Wave Field Synthesis (WFS)
 - Vector Base Amplitude Panning (VBAP)
 - Ambisonic Amplitude Panning (AAP)
 - Generic Renderer
 - Binaural Playback Renderer (BPB)
 - NFC-HOA Renderer → *coming soon!*
- runs on Linux and Mac OSX
- uses the *Jack Audio Connection Kit* (JACK)
- graphical user interface (Qt) and network interface (TCP/IP)
- Free and Open Source Software (GPLv3)
- <http://tu-berlin.de/?ssr>

The SoundScape Renderer (SSR).

Graphical User Interface.



Important Notes

- compile with optimization!
 - e.g. g++ -O3
- be aware of cache effects!
 - memory locality
 - false sharing
- look for bottlenecks with a profiler!
 - OProfile, gprof, ...

Conclusion

- goal of MimoProcessor: to be ...
 - unobtrusive
 - easy to use
 - re-usable in different contexts
 - easily extensible (e.g. by policy-based design)
- parallelization: simple yet effective
 - trade-off between effort (to use, to implement) and performance
 - significant gain in performance, e.g. for HOA renderer
- unit tests are included
 - using the *CATCH* framework
- well documented
 - *Doxygen* documentation also available at the website

Future Work

- Audio Processing Framework (APF)
 - include delay line and partitioned convolution
 - implement *PortAudio* policy
- SoundScape Renderer (SSR)
 - re-write core using the MimoProcessor
 - port all existing renderers
 - include the brand-new NFC-HOA renderer

Thank you very much for your attention!
Questions?

Website: <http://tu-berlin.de/?apf>
Blog: <http://audio.qu.tu-berlin.de>