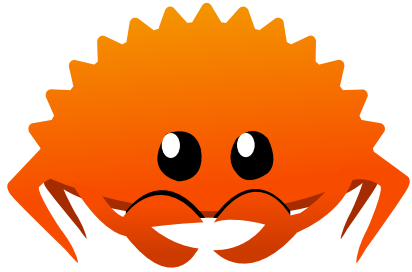


Experiences with Faust + Rust

F.A.U.S.T



whoami

- @obsoleszenz
- Techno DJ
- Tinkerer
- Exploring&Learning
- Means:
 - I don't know shit





My Projects

EQUIS

- DJ Mixer as Software
- Filters, filters, filters
- Inspired by Allen&Heath & MODEL1
- Implemented in FAUST + Rust

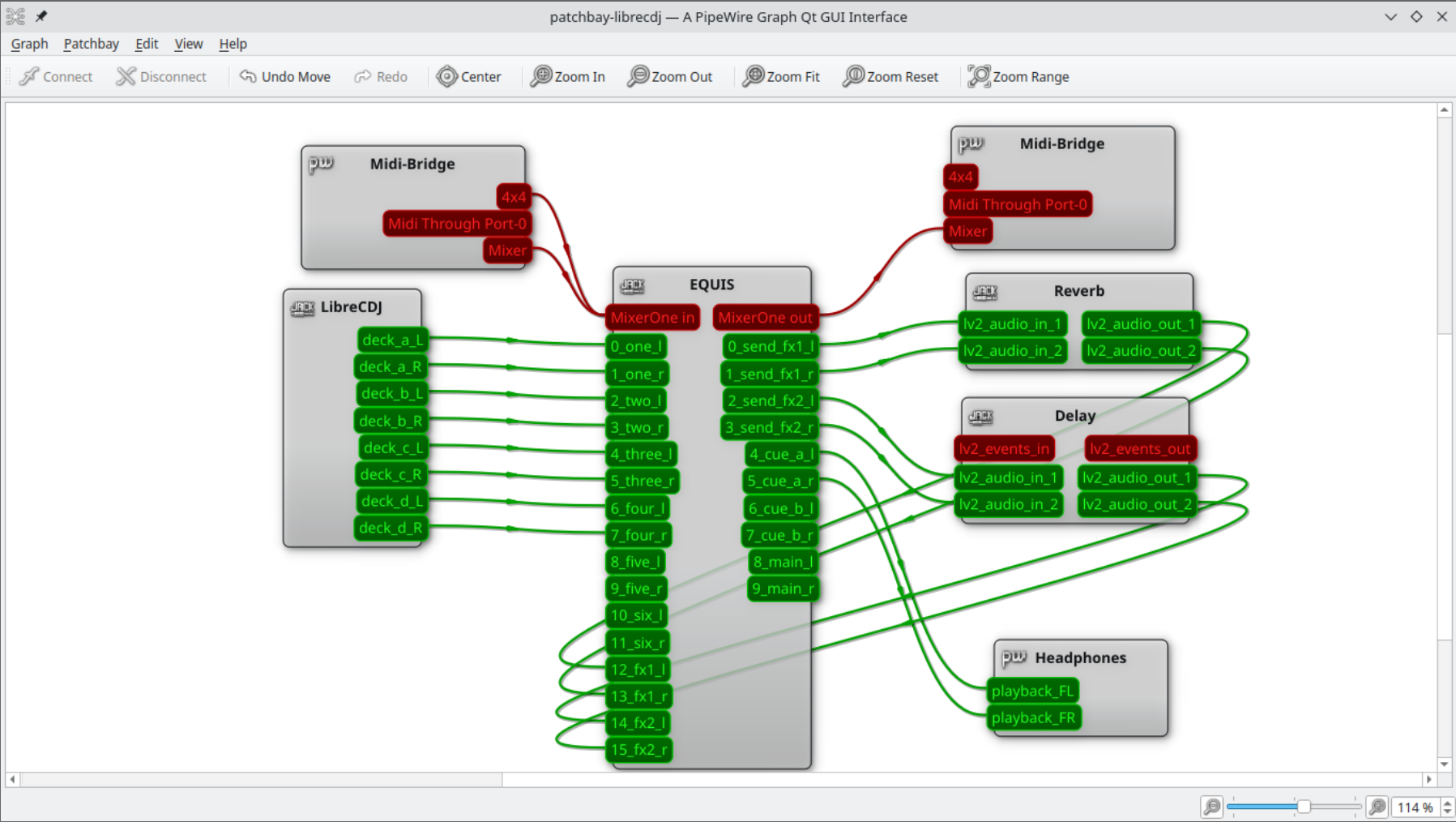
EQUIS

EQUIS
⌵ ⌶ ✕

Channel::One	Channel::Two	Channel::Three	Channel::Four	Channel::Five	Channel::Six	Channel::Fx1	Channel::Fx2
Gain: 0.50	Gain: 0.51	Gain: 0.50	Gain: 0.50	Gain: 0.50	Gain: 0.50	Gain: 0.40	Gain: 0.50
RMS: -70.00dB	RMS: -70.00dB	RMS: -70.00dB	RMS: -70.00dB	RMS: -70.00dB	RMS: -70.00dB	RMS: -70.00dB	RMS: -70.00dB
Peak: 3.51dB	Peak: 0.81dB	Peak: 0.00dB	Peak: 0.00dB	Peak: 0.00dB	Peak: 0.00dB	Peak: 0.00dB	Peak: 0.00dB
Volume: 1.00	Volume: 1.00	Volume: 0.00	Volume: 0.00	Volume: 1.00	Volume: 1.00	Volume: 1.00	Volume: 1.00
HighPass: 0.00	HighPass: 0.00	HighPass: 0.00	HighPass: 0.00	HighPass: 0.00	HighPass: 0.00	HighPass: 0.00	HighPass: 0.00
LowPass: 0.05	LowPass: 1.00	LowPass: 1.00	LowPass: 1.00	LowPass: 1.00	LowPass: 1.00	LowPass: 1.00	LowPass: 1.00
Sculpt Frequency: 0.32	Sculpt Frequency: 0.58	Sculpt Frequency: 0.50	Sculpt Frequency: 0.50	Sculpt Frequency: 0.50	Sculpt Frequency: 0.50	Sculpt Frequency: 0.50	Sculpt Frequency: 0.50
Sculpt Gain: 0.00	Sculpt Gain: 0.64	Sculpt Gain: 0.50	Sculpt Gain: 0.50	Sculpt Gain: 0.50	Sculpt Gain: 0.50	Sculpt Gain: 0.50	Sculpt Gain: 0.50
Send Fx1: 0.00	Send Fx1: 0.00	Send Fx1: 0.24	Send Fx1: 0.55	Send Fx1: 0.00	Send Fx1: 0.00	Send Fx1: 0.00	Send Fx1: 0.00
Send Fx2: 0.00	Send Fx2: 0.00	Send Fx2: 0.00	Send Fx2: 0.40	Send Fx2: 0.00	Send Fx2: 0.00	Send Fx2: 0.00	Send Fx2: 0.00
SubFilter: One	SubFilter: One	SubFilter: One	SubFilter: One	SubFilter: None	SubFilter: None	SubFilter: None	SubFilter: None
Cue A: On	Cue A: Off	Cue A: Off	Cue A: Off	Cue A: Off	Cue A: Off	Cue A: Off	Cue A: Off
Cue B: Off	Cue B: Off	Cue B: On	Cue B: Off	Cue B: Off	Cue B: Off	Cue B: Off	Cue B: Off

SubFilter::One	SubFilter::Two	Cue::A	Cue::B	Main
HP: On	HP: Off	Gain: 0.40	Gain: 0.50	Gain: 0.44
BP: Off	BP: Off	RMS: -70.00dB	RMS: -70.00dB	RMS: -70.00dB
LP: Off	LP: Off	Mix	Split L	
Resonance: 0.13	Resonance: 0.00	Mix Balance: 1.00	Bypass Limiter: Off	
Frequency: 0.15	Frequency: 0.00	Bypass Limiter: Off	Pre EQ: On	
Active: On	Active: Off	Pre EQ: Off		

EQUIS



EQUIS



LibreCDJ

- Also written in rust
- Free Open Source DJ Software
- Reliable & good sounding
- Club Ready™
- Fun to use
- Alternative to proprietary + expensive solutions

librecdj
128 BPM

File Edit **View** Help

Hide Library CTRL + Z
 Zoom Waveforms Space

Barbosa - Pain

+00:02:14 / 00:05:30
 143.00 [4m] / 145 {E5}
 techno dark rising boring..

1 2 3 4 5 6 7 8 A

Falling Echoes - Hypnotize..

+00:02:14 / 00:05:30
 143.00 [4m] / 145 {E5}
 techno dark rising boring..

1 2 3 4 5 6 7 8 B

DISCHARGED

+---:---:-- / ---:---:--
 1.0 [] / 0.0 { }

1 2 3 4 5 6 7 8 C

Barbosa - Pain

+00:02:14 / 00:05:30
 143.00 [4m] / 145 {E5}
 techno dark rising boring..

1 2 3 4 5 6 7 8 D

Techno > Playlists > test.m3u

	Some SubFolder	10	
	Falling Echoes Hypnotize Everything	143.00 4m	#techno #dark #rising #boring-end
	Yant Encode the Stick	134.00 12a	#hypnotic #bubbly

/ SOME SEARCH
COVR INFO BRWS

FAUST + Rust related projects

- Maintaining
 - faust-build
 - faust-types
- lowpass-lr4-faust-nih-plug

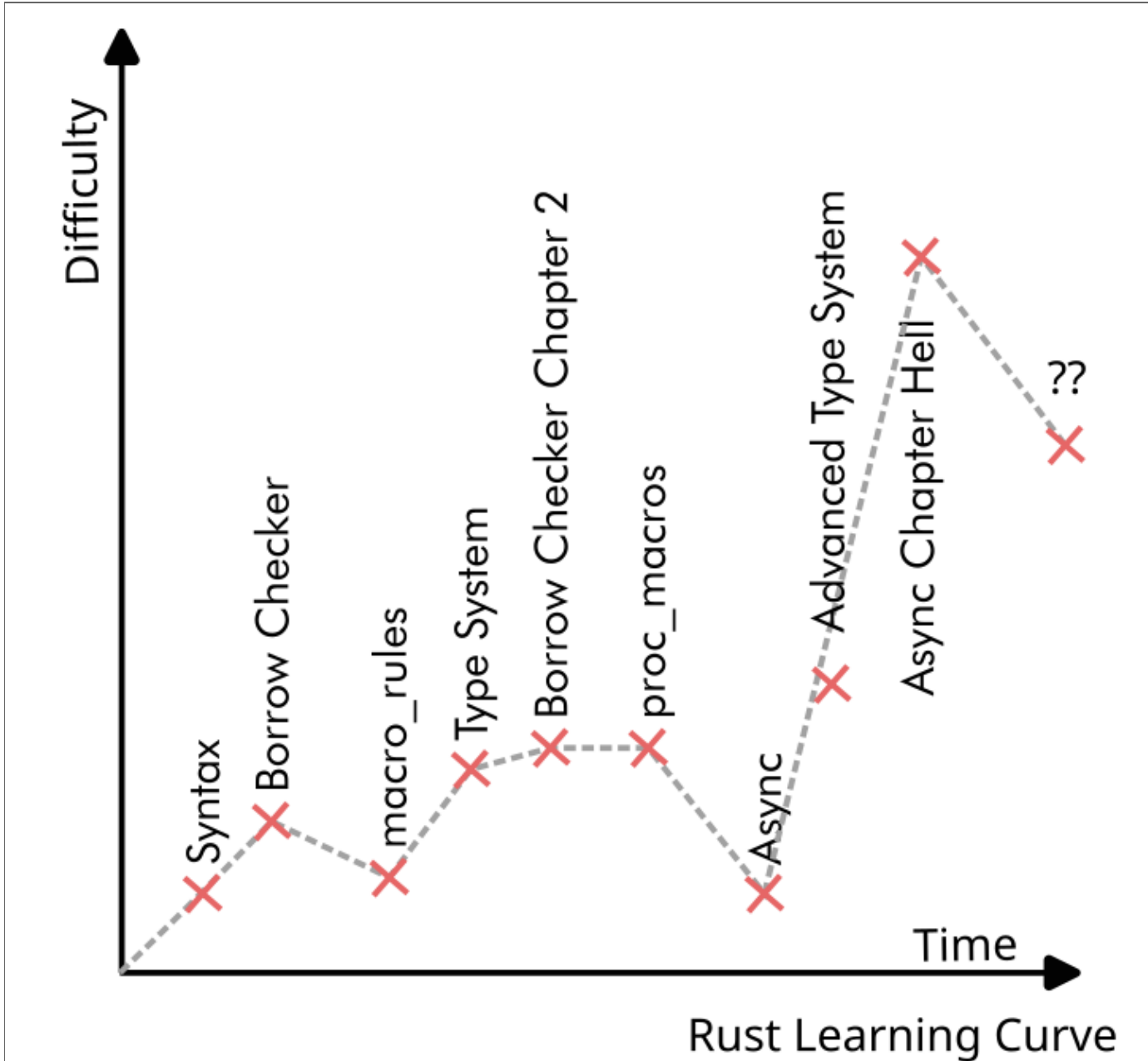
Sounds interesting? Join us!

We are looking for likeminded hackers, designers, musicians + friends
Find us on [codeberg](#)/[github](#) or send me an e-mail

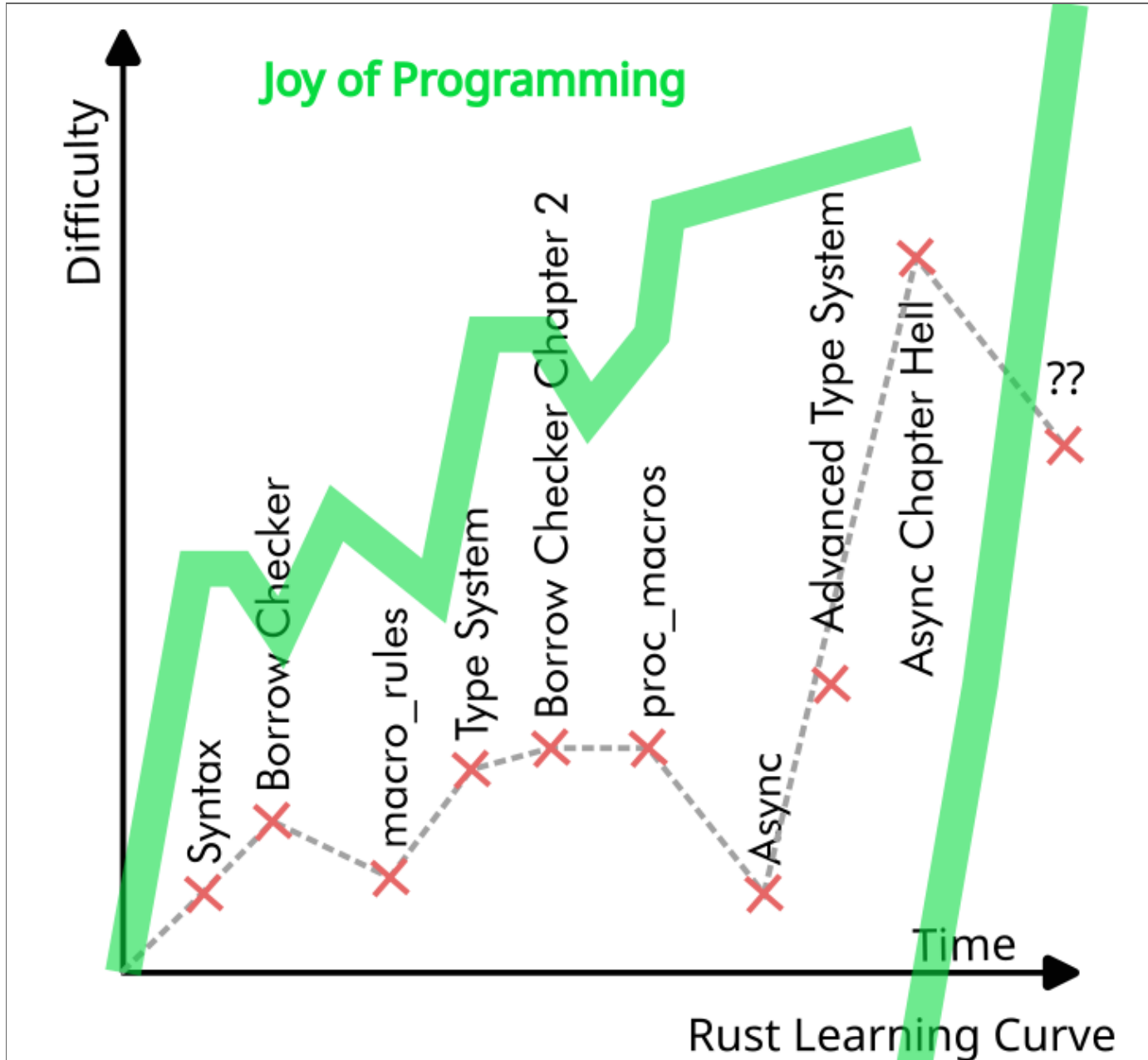
Links

- codeberg.org/obsoleszenz
- soundcloud.com/obsoleszenz
- @obsoleszenz@nerdculture.de
- obsoleszenz@riseup.net

Why NOT rust?



Why rust?



Why rust?

Helpful Compiler

Many sound Concepts

Beautiful Syntax

Good cross compilation

Error Handling

LISP like Macros Built-in Linter

Workspaces

Traits

Low Level

Semicolons!

`Option<T> gt null`

No garbage collector

Performance

Memory Safety

Pattern Matching

Data Ownership

Built-in Toolchain Management

Package Manager

Zero Cost Abstractions

Async

Algebraic Type System

Why rust?

- RUST is not a better C/C++ but a new way of programming

Why Faust?

- Beautiful syntax & concept
- Fast prototyping & exploration
- Reliable in production
- Good standard library & ecosystem

combine(faust, rust)


- Both progressive/functional languages
- Runs almost everywhere
 - Linux/FreeBSD/Mac/Win/Android/ios/arduino/vst/lv2/clap...
- Powerful modern system language + powerful DSP language
- Combining the Ecosystems

Let's build something!

volume

peak


8dB



reset


rms

-15.728d



volume

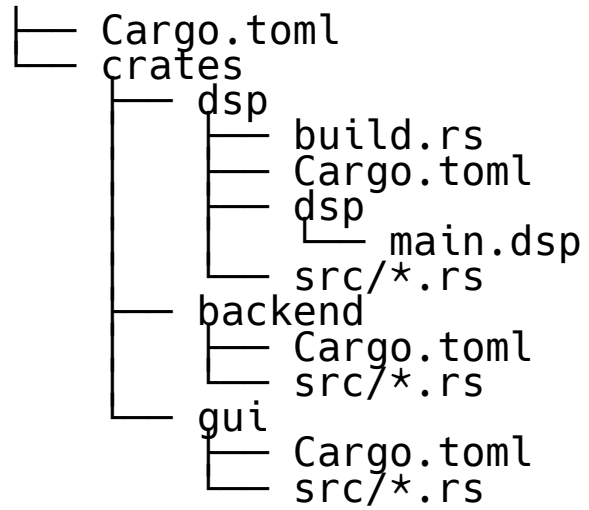
1



How do I do it?

- Dependencies
 - rust-faust ecosystem
 - jack
 - egui
- codeberg.org/obsoleszenz/rust-faust-jack-egui-template

Project Structure



DSP: Adding dependencies

- `cargo add faust-build faust-types`
- Checkout <https://github.com/Frando/rust-faust>

DSP: Add some faust

- Add faust code into dsp/dsp/main.dsp

```
import("stdfaust.lib");

p_volume = hslider("volume", 1.0, 0.0, 1.0, 0.01) : si.smoo;
p_peak_reset = button("reset");

holdMax(resetTrigger) = _ : (_,_ : resettableMax(resetTrigger)) ~ _
with {
    resettableMax(resetTrigger) = _,_ <: max,(!,_) :
select2(resetTrigger);
};

stereo2mono = _,_ :> abs : _ / 2 ;

mono_dbmeter(reset) = max(ba.db2linear(MIN_DB)) : ba.linear2db :
min(MAX_DB) : max ~ -(80.0/ma.SR) : holdMax(ba.pulse(HOLD_PEAK))
<:
    holdMax(reset)
    :
    hbargraph("rms[unit:dB]", MIN_DB, MAX_DB),
    hbargraph("peak[unit:dB]", MIN_DB, MAX_DB)
    :>

with {
    MIN_DB = -70;
    MAX_DB = 8;
    // window size for RMS
    WINDOW_SIZE = 100;
    // hold time for peak value in samples
    HOLD_PEAK = ma.SR/8;
};
```

```
dbmeter(reset) = <: _,_, stereo2mono : _, attach( _,  
mono_dbmeter(reset));  
process = par(i, 2, _ * p_volume) : dbmeter(p_peak_reset);
```

DSP: faust-build

crates/dsp/build.rs

```
fn main() {  
    #[cfg(feature = "faust-rebuild")]  
    faust_build::FaustBuilder::new("dsp/main.dsp", "src/faust_dsp.rs")  
        .set_struct_name("FaustDSP")  
        .build();  
}
```

DSP: faust-build

- Define faust-rebuild feature in crates/dsp/Cargo.toml
- ```
cargo -p dsp --features faust-rebuild
```
- This will generate crates/dsp/src/faust\_dsp.rs

# DSP: Wrapping

- generated rust code is not idiomatic
- until that's fixed we wrap it!
- Check out crop2000 prs & issues

- ```
pub struct DSP {  
    faust_dsp: FaustDSP,  
}  
  
impl DSP {  
    ...  
}
```

DSP: Wrapping Basics

```
impl DSP {
    pub fn new() -> Self {
        Self {
            faust_dsp: FaustDSP::new(),
        }
    }
    pub fn set_sample_rate(&mut self, sample_rate: usize) {
        self.faust_dsp.init(sample_rate as i32);
    }
    pub fn sample_rate(&self) -> usize {
        self.faust_dsp.get_sample_rate() as usize
    }
}
```

DSP: Wrapping Parameters

```
impl FaustDsp for FaustDSP {
  fn build_user_interface static(ui_interface: &mut dyn UI<Self::T>) {
    ui_interface.open_vertical_box("Volume");
    ui_interface.declare(Some(ParamIndex(0)), "unit", "dB");
    ui_interface.add_horizontal bargraph("peak", ParamIndex(0),
-7e+01, 8.0);
    ui_interface.add button("reset", ParamIndex(1));
    ui_interface.declare(Some(ParamIndex(2)), "unit", "dB");
    ui_interface.add_horizontal bargraph("rms", ParamIndex(2), -7e+01,
8.0);
    ui_interface.add_horizontal_slider("volume", ParamIndex(3), 1.0,
0.0, 1.0, 0.01);
    ui_interface.close_box();
  }
}
```


DSP: Wrapping Parameters

```
use faust_types::ParamIndex;

#[derive(Debug, PartialEq, Eq)]
pub enum Parameter {
    Volume,
    Reset,
}

impl From<Parameter> for ParamIndex {
    fn from(value: Parameter) -> ParamIndex {
        match value {
            Parameter::Volume => ParamIndex(3),
            Parameter::Reset => ParamIndex(1),
        }
    }
}
```

DSP: Wrapping UI Parameters

```
pub enum UIParameter {  
    Rms,  
    Peak,  
}  
  
impl From<UIParameter> for ParamIndex {  
    fn from(value: UIParameter) -> ParamIndex {  
        match value {  
            UIParameter::Rms => ParamIndex(2),  
            UIParameter::Peak => ParamIndex(0),  
        }  
    }  
}
```

DSP: Wrapping Parameters

- ```
impl DSP {
 pub fn set_parameter(&mut self, parameter: Parameter, value: f32) {
 self.faust_dsp.set_param(parameter.into(), value);
 }

 pub fn parameter(&mut self, parameter: Parameter) -> f32 {
 self.faust_dsp
 .get_param(parameter.into())
 .expect("Invalid parameter index")
 }

 pub fn ui_parameter(&self, ui_parameter: UIParameter) -> f32 {
 self.faust_dsp
 .get_param(ui_parameter.into())
 .expect("Invalid parameter index")
 }
}
```

## DSP: Problems with Faust compute

- `inputs: &[&[f32]]`
- Compiler cannot verify input/output buffer length
- A slice of slices is annoying to use
- count type is i32 instead of usize

## DSP: Wrapping compute

```
impl DSP {
 pub const INPUTS: usize = 2;
 pub const OUTPUTS: usize = 2;

 pub fn compute(
 &mut self,
 count: usize,
 inputs: &[I; Self::INPUTS],
 outputs: &mut [O; Self::OUTPUTS],
) where
 I: AsRef<[f32]>,
 O: AsMut<[f32]>,
 {
 // We need to change the memory layout of inputs/outputs so
 // that FaustDSP likes it.
 // As this is a bit ugly I hope we can move this to the faust
 // generated code at some point.
 let inputs: [&[f32]; Self::INPUTS] = [
 inputs[0].as_ref(),
 inputs[1].as_ref(),
];

 let mut outputs_iter = outputs.iter_mut();
 let mut outputs: [&mut [f32]; Self::OUTPUTS] = [
 outputs_iter.next().unwrap().as_mut(),
 outputs_iter.next().unwrap().as_mut(),
];

 self.faust_dsp.compute(
 count as i32,
 inputs.as_slice(),
 outputs.as_mut_slice()
);
 }
}
```

## DSP: State

```
use crate::{parameter::UIParameter, Parameter, DSP};

#[derive(Debug, Clone, PartialEq, Default)]
pub struct DSPState {
 pub rms: f32,
 pub peak: f32,
 pub p_volume: f32,
 pub p_reset: f32,
}

impl DSPState {
 pub fn update(&mut self, dsp: &mut DSP) {
 self.rms = dsp.ui_parameter(UIParameter::Rms);
 self.peak = dsp.ui_parameter(UIParameter::Peak);
 self.p_volume = dsp.parameter(Parameter::Volume);
 self.p_reset = dsp.parameter(Parameter::Reset);
 }
}
```

## DSP: State

```
impl DSP {
 pub fn update_state(&mut self, dsp_state: &mut DSPState) {
 dsp_state.update(self)
 }
}
```

## DSP: Usage examples

```
let mut dsp = DSP::new();
let dsp.set_sample_rate(44_800);

// Example with arrays
const MAX_BUFFER: usize = 1024;
let inputs = [[0.0_f32; MAX_BUFFER]; DSP::inputs()];
let mut outputs = [[0.0_f32; MAX_BUFFER]; DSP::outputs()];
dsp.compute(128, &inputs, &mut outputs);

// Example with resizable vectors
let max_buffer_size = 1024;
let mut inputs = vec![vec![0_f32; max_buffer_size]; DSP::INPUTS];
let mut outputs = vec![vec![0_f32; max_buffer_size]; DSP::OUTPUTS];
dsp.compute(
 128,
 inputs.as_slice().try_into().unwrap(),
 outputs.as_mut_slice().try_into().unwrap()
);

// Setting parameters
dsp.set_parameter(Parameter::Volume, 1.0);
dsp.set_parameter(Parameter::Reset, 1.0);

// Reading ui parameters
let rms = dsp.get_ui_parameter(UIParameter::Rms);
let peak = dsp.get_ui_parameter(UIParameter::Peak);

// Using DspState
let mut state = DspState::default();
dsp.update_state(&mut dsp_state);
```



Anyone programming in C?

# Welsh or C standard library function?

**mbsrtowcs**

**rhowch**

**strxfrm**

**cwtch**

**mwyn**

**wcstold**

**wmffre**

**wcsoll**



Anyone programming in C?

**Welsh** or C standard  
library function?

**mbsrtowcs**

**rhowch**

**strxfrm**

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## Backend

- Should initiate jack
- Be the bridge between ui and audio thread
- Receive parameter changes
- Send state updates

## Backend: JackBackend

- `cargo add jack rtrb triple_buffer`
- ```
pub struct JackBackend {  
    dsp: DSP,  
    jack: jack::Client,  
    ports_audio_in: Vec<Port<AudioIn>>,  
    ports_audio_out: Vec<Port<AudioOut>>,  
    state_writer: StateWriter,  
    command_consumer: CommandConsumer,  
}
```

Backend: Types

- ```
use rtrb::{Consumer, Producer};
use triple_buffer::{Input, Output, TripleBuffer};
pub type StateReader = Output<DSPState>;
pub type StateWriter = Input<DSPState>;

pub type Command = (Parameter, f32);
pub type CommandProducer = Producer<Command>;
pub type CommandConsumer = Consumer<Command>;
```

## Backend: JackBackend::new()

```
impl JackBackend {
 pub fn new() -> (Self, StateReader, CommandProducer) {
 let dsp = DSP::new();

 let (jack, status) =
 jack::Client::new("IFC24",
 jack::ClientOptions::NO_START_SERVER).unwrap();
 let mut ports_audio_in: Vec<Port<AudioIn>> = Vec::new();
 let mut ports_audio_out: Vec<Port<AudioOut>> = Vec::new();

 ports_audio_in.push(
 jack.register_port("in_l", jack::AudioIn)
 .expect("Failed registering in_l port"),
);
 ports_audio_in.push(
 jack.register_port("in_r", jack::AudioIn)
 .expect("Failed registering in_r port"),
);
 ports_audio_out.push(
 jack.register_port("out_l", jack::AudioOut)
 .expect("Failed registering out_l port"),
);
 ports_audio_out.push(
 jack.register_port("out_r", jack::AudioOut)
 .expect("Failed registering out_r port"),
);

 let (state_writer, state_reader) =
 TripleBuffer::new(&DSPState::default()).split();
 let (command_producer, command_consumer) =
 rtrb::RingBuffer::new(24);

 (
 Self {
 dsp,
 jack,
 ports_audio_in,
```



```
 ports_audio_out,
 state_writer,
 command_consumer,
 },
 state_reader,
 command_producer,
)
}
}
```

## Backend: JackBackend::run()

```
impl JackBackend {
 pub fn run(mut self) -> JackJoinHandle {
 let (shutdown_send, shutdown_recv) = sync_channel::<()>(1);
 let join_handle = std::thread::spawn(move || {
 // init dsp with a given sample rate
 let mut sample_rate = self.jack.sample_rate();
 self.dsp.set_sample_rate(sample_rate);

 // init input and output buffers
 let mut max_buffer_size = (self.jack.buffer_size() * 2) as
usize;
 let mut inputs = vec![vec![0_f32; max_buffer_size];
DSP::INPUTS];
 let mut outputs = vec![vec![0_f32; max_buffer_size];
DSP::OUTPUTS];
 info!("start jackbackend with
max_buffer_size={max_buffer_size} and sample_rate={sample_rate}");

 let mut last_ui_state_update = Instant::now()
 .checked_sub(MAX_ELAPSED_UI_WRITE * 2)
 .unwrap();

 // create jack process closure that runs for each buffer
 let process_callback = move |jack_client: &jack::Client,
ps: &jack::ProcessScope|
 -> jack::Control {
 let mut state_changed = false;

 let len: usize = ps
 .n_frames()
 .try_into()
 .expect("can't cast jack n_frames to usize");

 // Handle jack requesting a bigger buffer as we can handle
 if len > max_buffer_size {
 error!(
 "jack wants {} samples but our buffer can only
```

```

hold {}. resizing buffers. this might cause xruns.",
 len, max_buffer_size
);
 max_buffer_size = len * 2;
 inputs = vec![vec![0_f32; max_buffer_size];
DSP::INPUTS];
 outputs = vec![vec![0_f32; max_buffer_size];
DSP::OUTPUTS];
}

let jack_sample_rate = jack_client.sample_rate();
// Handle jack changing the sample rate
if jack_sample_rate != sample_rate {
{jack_sample_rate}");
 self.dsp.set_sample_rate(jack_sample_rate);
 sample_rate = jack_sample_rate;
}

// copy audio input for all ports from jack to the faust
input buffer
inputs
 .iter_mut()
 .enumerate()
 .for_each(|(index_port, input)| {
 let port_audio_in =
self.ports_audio_in[index_port].as_slice(ps);
 input[0..len].copy_from_slice(port_audio_in);
 });

// Apply all commands the backend sent us
while let Ok(command) = self.command_consumer.pop() {
 self.dsp.set_parameter(command.0, command.1);
 state_changed = true;
}

// Flush denormals and then run the faust dsp
no_denormals(|| {
 self.dsp.compute(
 len,
 inputs
 .as_slice()
 .try_into()
 .expect("Invalid amount of inputs channels"),
 outputs

```

```

 .as_mut_slice()
 .try_into()
 .expect("Invalid amount of output channels"),
 });
});
// copy audio output for all ports from faust to the jack
output
outputs
 .iter_mut()
 .enumerate()
 .for_each(|(index_port, output)| {
 let port =
self.ports_audio_out[index_port].as_mut_slice(ps);
 port.copy_from_slice(&output[0..len]);
 });

// Update state if needed
if state_changed || last_ui_state_update.elapsed() >
MAX_ELAPSED_UI_WRITE {
 let dsp_state = self.state_writer.input_buffer();
 self.dsp.update_state(dsp_state);
 self.state_writer.publish();
 last_ui_state_update = Instant::now();
}

jack::Control::Continue
};

// init jack process handler.
let process =
jack::ClosureProcessHandler::new(process_callback);

// activate the jack, which starts the processing.
let active_client = jack::AsyncClient::new(self.jack, (),
process).unwrap();

shutdown_recv
 .recv()
 .expect("error on shutdown_recv channel");

active_client
 .deactivate()
 .expect("failed deactivating jack client");

```

```
 debug!("stopped jack backend & thread");
 });
 JackJoinHandle {
 join_handle: Some(join_handle),
 shutdown_send,
 }
}
}
```

## Backend: Backend

```
use dsp::Parameter;
use tracing::error;

use crate::{
 jack::{CommandProducer, JackJoinHandle, StateReader},
 JackBackend,
};

pub struct Backend {
 jack_join_handle: Option<JackJoinHandle>,
 command_producer: CommandProducer,
}

impl Backend {
 pub fn new() -> (Self, StateReader) {
 let (jack_backend, state_reader, mut command_producer) =
 JackBackend::new();
 let jack_join_handle = jack_backend.run();
 (
 Self {
 jack_join_handle: Some(jack_join_handle),
 command_producer,
 },
 state_reader,
)
 }

 pub fn command(&mut self, parameter: Parameter, value: f32) {
 if let Err(err) = self.command_producer.push((parameter, value)) {
 error!("Failed sending command to backend because of
err={err:?}");
 }
 }
}

impl Drop for Backend {
 fn drop(&mut self) {
```

```
 if let Some(jack_join_handle) = self.jack_join_handle.take() {
 jack_join_handle.shutdown();
 }
}
```

## Check-In

**My kidnappers returning me  
after listening to me talk  
about rust for two hours**







# Gui: app.rs

```
use std::time::Duration;
use backend::{Backend, Parameter, StateReader};
use eframe::egui::{self, Button, Sense};

const REFRESH_RATE_HZ: f32 = 30.0;

pub struct App {
 backend: Backend,
 state_reader: StateReader,
}

impl App {
 /// Called once before the first frame.
 pub fn new(_cc: &eframe::CreationContext<'>) -> Self {
 let (backend, state_reader) = Backend::new();

 Self {
 backend,
 state_reader,
 }
 }
}

impl eframe::App for App {
 /// Called each time the UI needs repainting, which may be many times
 per second.
 fn update(&mut self, ctx: &egui::Context, _frame: &mut eframe::Frame)
 {
 let state = self.state_reader.read();

 egui::CentralPanel::default().show(ctx, |ui| {
 // The central panel the region left after adding TopPanel's
 and SidePanel's
 ui.heading("IFC24 Example");

 ui.add(
```

```

 equi::::from_get_set(0.0..=1.0, |value| {
 if let Some(value) = value {
 self.backend.command(Parameter::Volume, value as
f32);
 return value;
 }
 }
 state.p_volume as f64
 })
 .text("Volume"),
);
ui.add(equi::Label::new(format!("RMS: {}dB", state.rms)));
ui.horizontal(|ui| {
 ui.add(equi::Label::new(format!("Peak: {}dB",
state.peak)));
 let reset_button_response =
ui.add(Button::new("Reset").sense(Sense::drag()));
 if reset_button_response.drag_started() {
 self.backend.command(Parameter::Reset, 1.0);
 }
 if reset_button_response.drag_released() {
 self.backend.command(Parameter::Reset, 0.0);
 }
});
});
ctx.request_repaint_after(Duration::from_secs_f32(1.0 /
REFRESH_RATE_HZ));
}
}

```

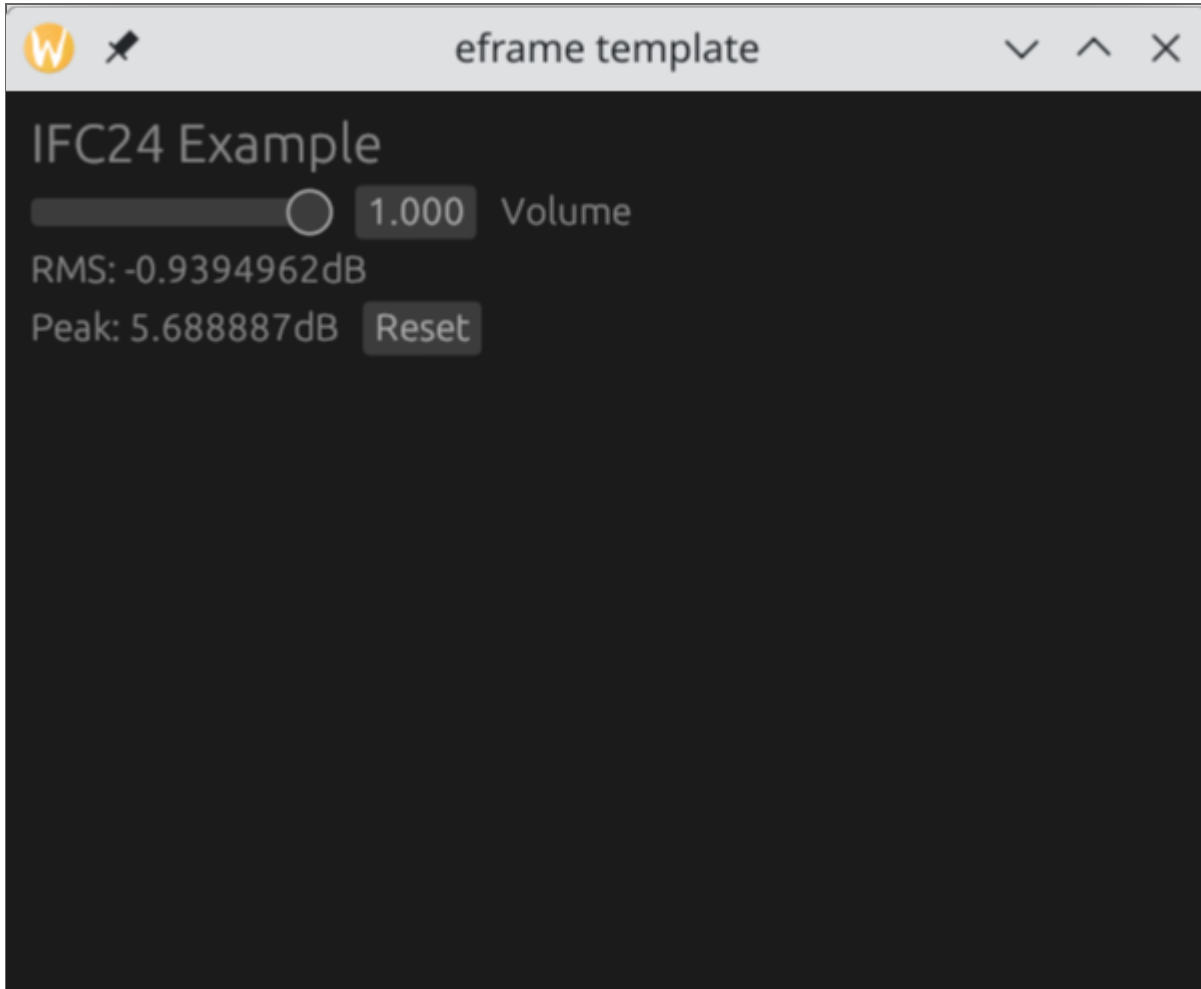
## Gui: main.rs

```
mod app;

use app::App;
use eframe::egui;

fn main() -> eframe::Result<()> {
 tracing_subscriber::fmt::init();
 let native_options = eframe::NativeOptions {
 viewport: egui::ViewportBuilder::default()
 .with_inner_size([400.0, 300.0])
 .with_min_inner_size([300.0, 220.0]),
 ..Default::default()
 };
 eframe::run_native(
 "eframe-template",
 native_options,
 Box::new(|cc| Box::new(App::new(cc))),
)
}
```

Tadaaaa!



The screenshot shows a window titled "eiframe template" with a standard OS title bar. The main content area has a dark background and displays the following information:

- Title: IFC24 Example
- Volume control: A slider bar with a circular knob, a text box containing "1.000", and the label "Volume".
- RMS: -0.9394962dB
- Peak: 5.688887dB
- A "Reset" button.

## Recommendations

- nih\_plug
- vizia
- slint
- symphonia



Speaker notes