



~Pure Data~

Normalizing the signal

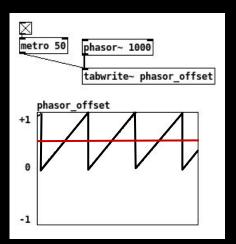
Goal: Obtaining the best dynamic range by fitting the gain of the signal into certain ranges.

Two step-process:

- Removing the DC Offset
- Normalizing the signal

What is DC Offset?
Mean amplitude displacement from 0.

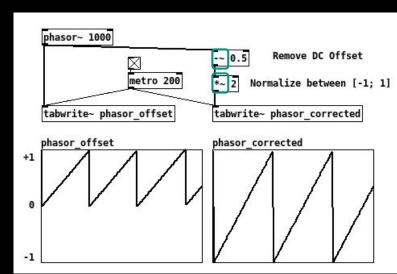
What is Normalization?
Adjust the gain to peak at the maximum the sound card allows before clipping: [-1, 1].



How to normalize this signal?

DC Offset = 0.5

Don't forget it's a ~signal



Normalizing the signal

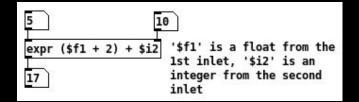
Example: from a Phasor to a Square

How to convert a sawtooth into a square?

output = 1 if signal > 0.5

0 otherwise

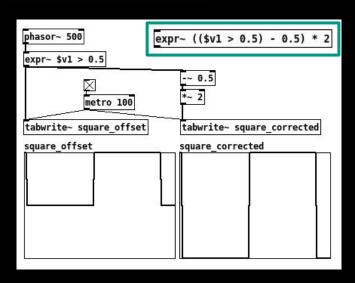
→ Introducing [expr] & [expr~]



The first inlet of [expr~] needs to be of type '\$v1' for a signal

Select 'Polygon' in the properties of the array



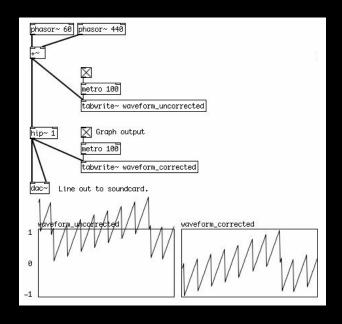


Normalizing the signal

Other way to get rid from the DC Offset

DC Offset = Direct Current Offset

- Continuous component that shifts the signal toward + or -
- As it is continuous, it has aO Hz frequency
- → A Hi-Pass filter with a very low cutoff frequency (below the perception) can remove it without altering the signal.



→ Put a Hi-Pass before each channel of your dac~

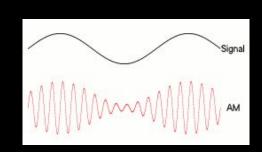
This won't normalize your signal:

→ Does not prevent clipping!

Amplitude Modulation

Earliest method to transmit audio in radio broadcast

<u>Definition</u>: varying the amplitude of a high frequency signal, the carrier signal, as a function of a lower frequency signal, the modulating signal (commonly the one containing the information to be transmitted).





Carrier signal high frequency

 $\overline{x_p}(t) = A_p \cos(\omega_p t)$



Modulating signal low frequency

 $\overline{x_m(t)} = A_m \cos(\omega_m t)$

Output: $y(t) = x_p(t) + kx_p(t)x_m(t)$

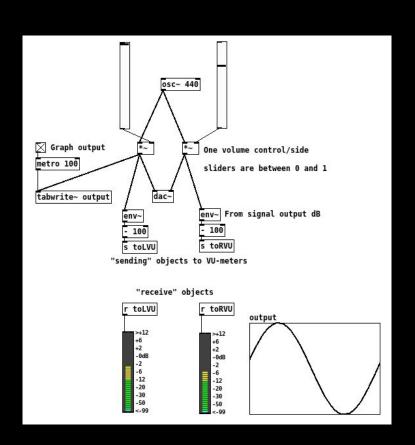
Amplitude Modulation Synthesis

Stereo Oscillator

- This patch is stereo: how would you do this?
- We want to visualize the output dB: which object?

Introducing the wireless connexion in Pd:

- → Use a "sending" object and a "receive" object formalized as [s *] and [r *]
- To use the VU-meter we need [env~] which take a signal and output its RMS amplitude in dB.
- → *First step*: building the stereo oscillator



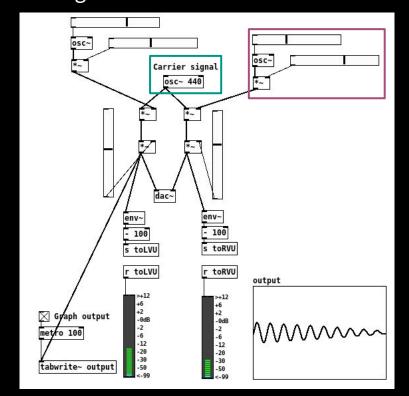
Amplitude Modulation Synthesis

Using one oscillator to modulate the gain of an other one

According to your intuition, how would you patch an AM?

[0: 10] Hertz Carrier signal Modulating signal osc~ 2 Depth [0;1] osc~ 440 That's a LFO (low-frequency osc below 20 Hz) dac~ 100 toRVU toLVU toLVU toRVU output Graph output

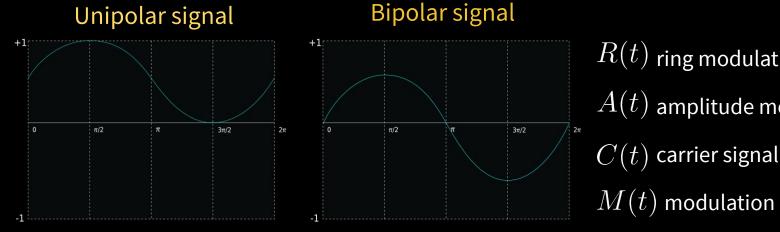
 Let's create a tremolo from this: adding AM on both sides



Ring Modulation

DC offset

Difference between AM & RM



R(t) ring modulation output

A(t) amplitude modulation output

M(t) modulation signal

RM: multiplication of two bipolar signals by each others $R(t) = C(t) \times M(t)$

Mean amplitude around 0

- The frequency of the carrier signal is not present in the resulting sound.
 - AM: M is a unipolar modulator, typically between 0 and 1. $A(t) = C(t) \times (M(t) + 1)$
- The carrier frequency is preserved.

Ring Modulation

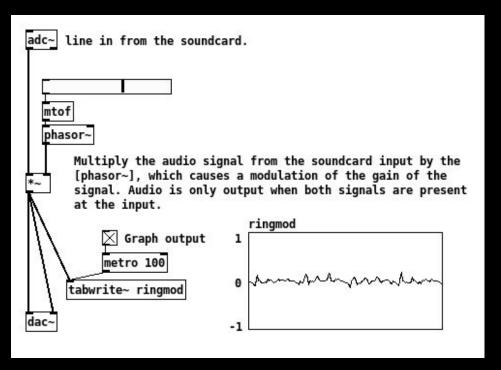
Making Alien's voice with ring modulation

Doctor Who - Cyberman voice



Which object will we use to catch our voice?

→ [adc~]



Frequency Modulation

By John Chowning



Stria (1977)

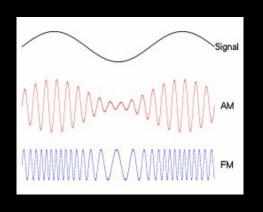
<u>Definition</u>: the information contained in the modulating signal is carried by varying the frequency of the carrier signal.

- Generally more robust than AM to transmit messages (less noise).
- Instable compare to AM regarding synthesis.
- Gives "natural" (& beautiful) sounds.

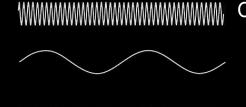
Mathematical intuition: sinusoid modulated by another sinusoid.

With: f_c carrier frequency, f_m modulating frequency and I_m modulation index, then:

output: $y(t) = \sin(2\pi f_c t + I_m \sin(2\pi f_m t))$



Difference between AM and FM



Carrier signal f_c

Modulating signal f_{m}



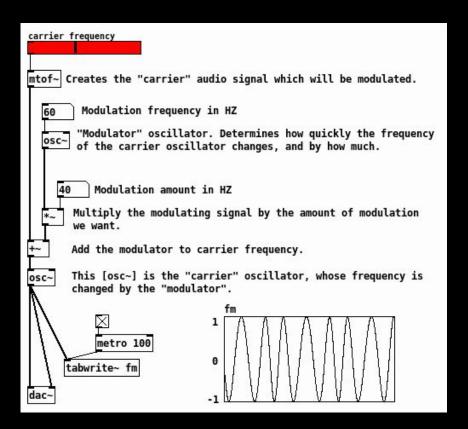
Output

Frequency Modulation

Simple fm patch

You can add colors on your objects to make your patch simpler

- For a very small amount of modulation: vibrato
- For a greater amount of modulation: glissando, or sweeping.



Step Sequencer

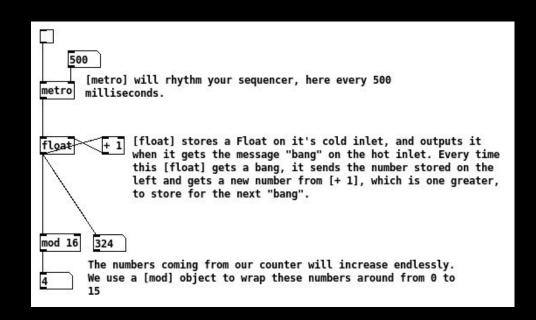
First Step: the counter

<u>Definition</u>: MIDI-based tool that divides a measure of music into a predetermined number of note value called steps.

→ We first need a counter!

How would you do it? We want 16 step sequencer

- → You will probably need [mod *] which wrap number around given value.
- → we want as output an increasing number modulo 16.



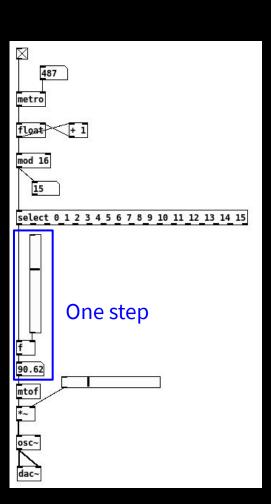
Step Sequencer

Second Step: the step sequencer

We will trigger a bang step by step using [select *] which compare numbers and send a bang if matching to the message.

When you are happy with one of your step, just copy paste 15 times

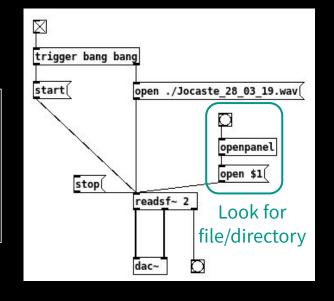
You can change osc, add enveloppes, configure your patch so that it plays harmonically, plays samples instead of notes...



Read a sound file

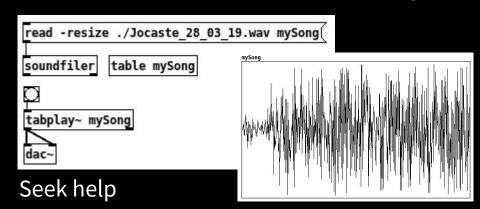
Reminder: Common sound file formats:

- No compression: .wav or .aiff
- Compression but no quality loss: .flac
- Compression and quality loss: .mp3
- → Pd objects depending on the format [readsf~] for .wav
- other formats not included in Pd Vanilla
- Don't use space or special characters for the name of your sound files
- Try to put your music in the same folder as your patch



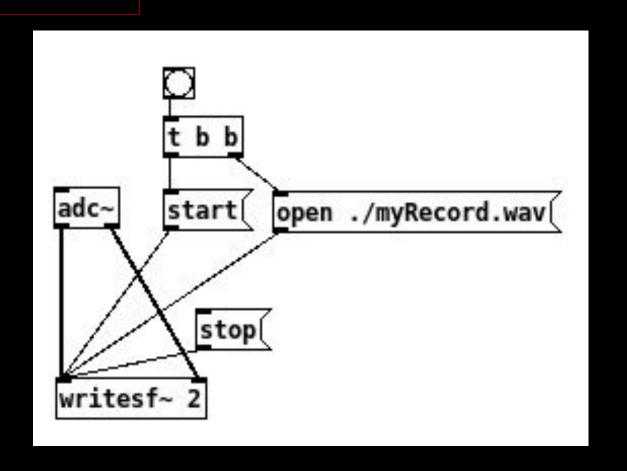
How would you loop the song?

[read[and [soundfiler] to "edit" the song



Write a sound file

How to record in Pd?



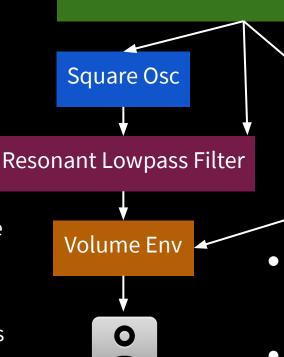
Kick Drum ♥

General Approach

 Percussive reverb with a white noise

Noise

- Square wave oscillator whose pitch is controlled by a pitch envelope.
- The volume envelope permits to have "note" and not continuous sound.



Pitch Env

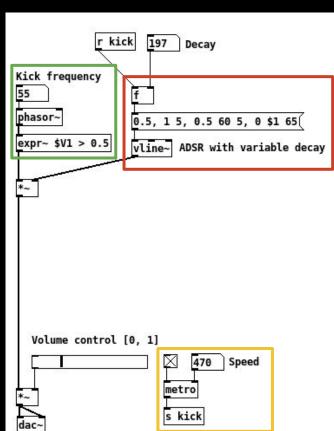
The cutoff freq of the lowpass is also driven by the pitch envelope

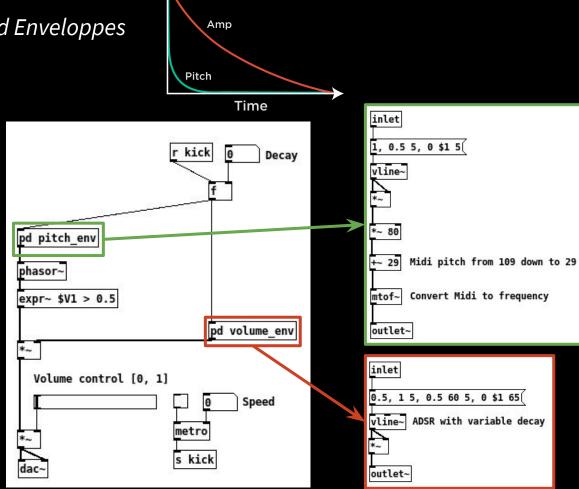
 Adding a sine driven by the pitch envelope

Sine Osc



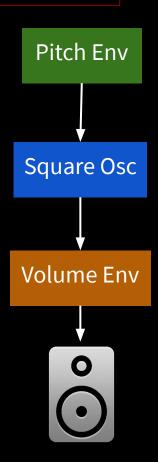
Part 1: Phasor and Enveloppes





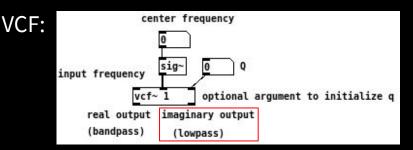


First step achievements



Next: adding a resonant low-pass filter, cutoff driven by our pitch enveloppe

→ Which object do we need?

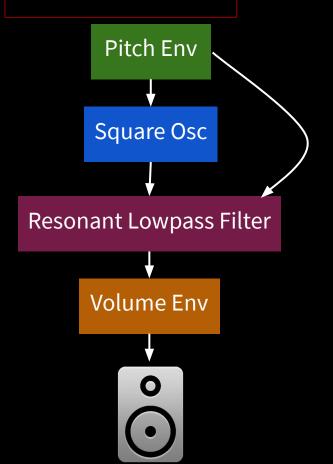


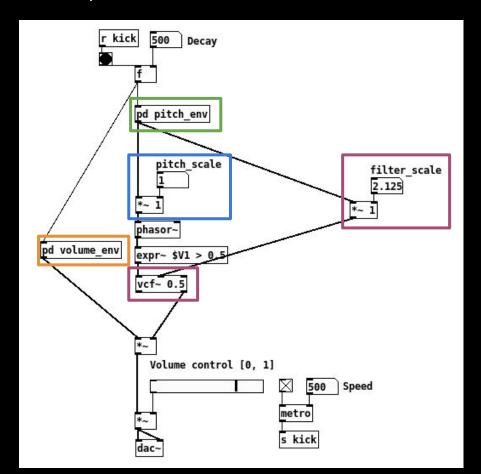
Where will we plug our pitch env?

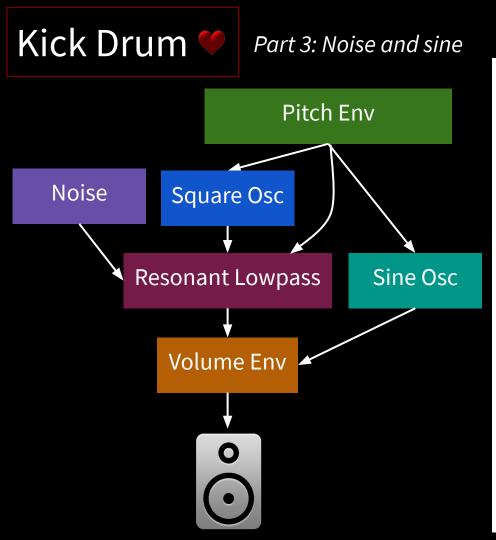
First, let's add a pitch scale, then the vcf with a filter scale.

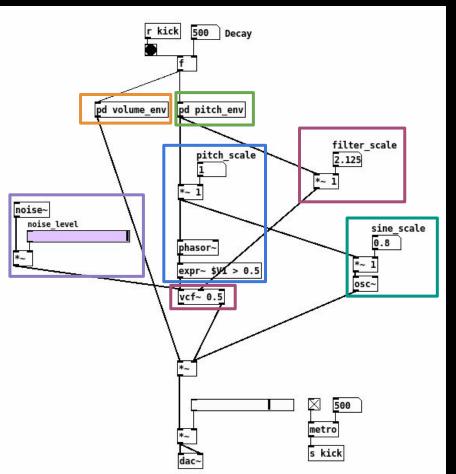
Kick Drum 💙

Part 2: Resonant low-pass filter and scales control





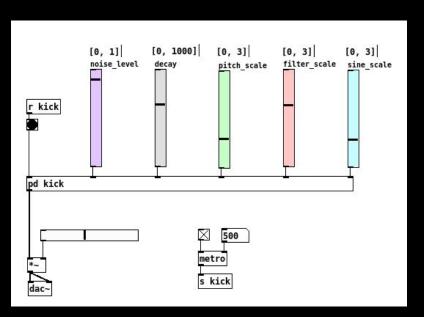


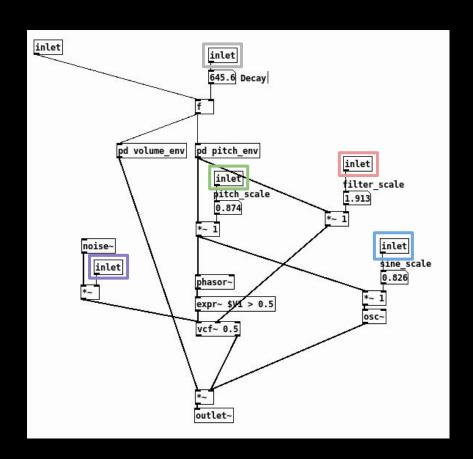


Kick Drum Part 4: Clean it up

How to simply subpatch:

- put your inlets and outlets appropriately
- cut your boxes with them
- plug respectfully with the order



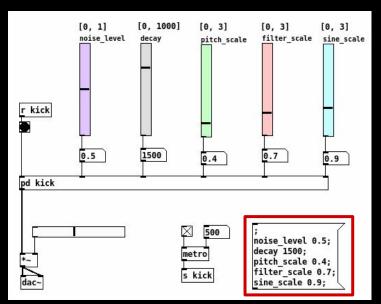




Part 4: Add preset

We will use messages to build presets.

- First add to all the concern sliders a label in "receive symbol"
- Then write a message beginning with;
 followed by all the label you want to drive

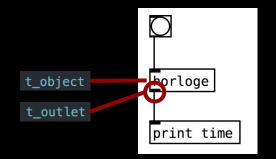




- So what is going on **inside** a given box ?
- We can even go deeper in Pure Data objects
- Possibility to define your own boxes:)
- The overall system defines **PD** externals
- Here we will code in C but still talk about objects
- PD provides a set of includes and specs
- Simple SDK with a clear notation
- Entirely dynamic linking / Runtime class loading
- Everything defined as a C struct
- Then simply a set of functions

- L. We need to **include the** PD header definitions
- 2. Then define the class of our object
 - This object reference is mandatory (cf. later)
 - . Need to manually define inlets and outlets

All objects have a default left-most hot inlet



Here we want to code a simple object
=> bang prints time

```
#ifndef HORLOGE H
# define HORLOGE H
                                              We need to include the PD header definitions
# include "m_pd.h"
                                              Then define the class of our object
                 *horloge_class;
static t class
typedef struct
                 _horloge
                                              This object reference is mandatory (cf. later)
                  x obj:
   t object
   t outlet
                  *h_out;
                                              Need to manually define inlets and outlets
t_horloge;
                                                            3 minimal functions to code
                 horloge_setup(void);
                                                          What happens at runtime (once)
void
                                                          Define object creation (add box)
                 *horloge_new(void);
void
                                                     3.
                                                          One method per message
void
                 horloge bang(t horloge *x);
                                                    (here we code what happens when a bang is received)
#endif
```

1. Runtime function (*_setup)

Reminder of the data structure

```
Class creation method class_new
```

Add the behavior for bang with class_addbang
Later we will also use class addmethod (messages)



2. Box creation function (*_new)

Reminder of the data structure

Return the created object



3. Message handling function (*_bang)

Reminder of the data structure

Specific object instance

```
void horloge_bang(t_horloge *x)

time_t rawtime;

struct tm *timeinfo;

time(&rawtime);

timeinfo = localtime(&rawtime);

outlet_symbol(x->h_out, gensym(asctime(timeinfo)));
}
```

Need to write symbols to a given symbol table

Write information to a specific outlet