

# Pure Data

Intermediate course



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Made with love for ATIAM

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# NORMALIZATION

Adjusting the dynamics range

# 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

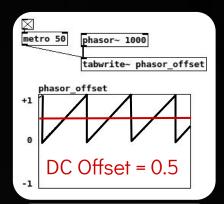
#### What is DC Offset?

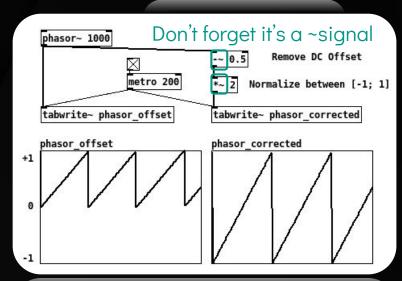
Mean amplitude displacement from 0.

#### What is Normalization?

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

How to normalize this signal?





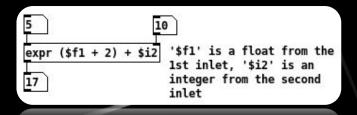
### Normalizing

How to convert a sawtooth into a square?

output = 1 if signal > 0.5 0 otherwise

Introducing [expr] & [expr~]

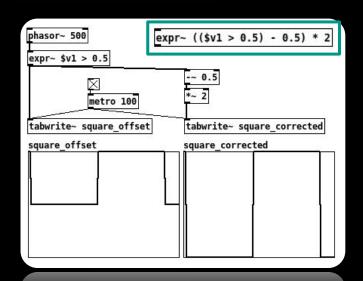
The first inlet of [expr~] needs to be of type '\$v1' for a signal



Select 'Polygon' in the properties of the array

Example: from a Phasor to a Square





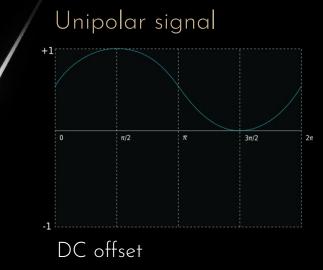
# 02 Modulations

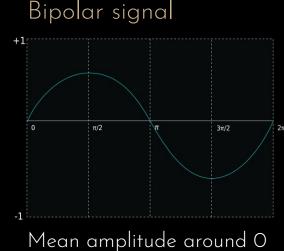
Amplitude, Ring and Frequency Modulations

#### MODULATION Difference between AM & RM

Multiplying audio signal

Modulation is typical in synthesis as it enriches the character of the sound and adds variance in tibre over time.





A(t) amplitude modulation outp C(t) carrier signal M(t) modulation signal

 $R(t) = C(t) \times M(t)$ 

R(t) ring modulation output

- Ring Modulation: multiplication of two bipolar signals by each others
- The frequency of the carrier signal is not present in the resulting sound.
- Amplitude Modulation: M is a unipolar modulator, typically between 0 and 1.
- lacktriangledown The carrier frequency is preserved.  $A(t) = C(t) \times (M(t) + 1)$

## RING MODULATION

$$R(t) = C(t) \times M(t)$$

We multiply 2 bipolar signals by each other resulting:

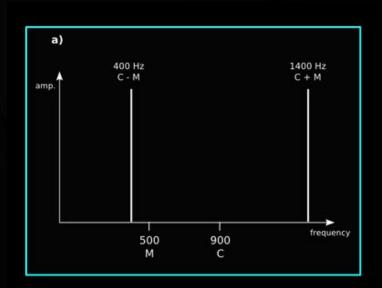
$$cos(\alpha n + \phi)cos(\beta n + \xi) = \frac{1}{2}(cos(\alpha + \beta)n + (\phi + \xi)) + cos(\alpha - \beta)n + (\phi - \xi))$$

We obtain two partials, one at the sum of the two original frequencies and one at their difference.

This shifts the component frequencies of a sound

Doctor Who
Cyberman voice





### STEREO PATCH

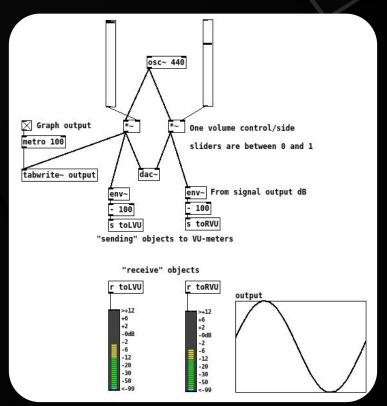
- This patch is stereo: how would you do this?
- We want to visualize the output dB: which object?

Introducing the wireless connexion in Pd:

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

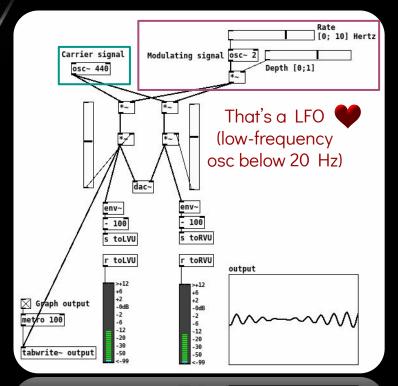
*First step*: build the stereo oscillator

#### Stereo Oscillator



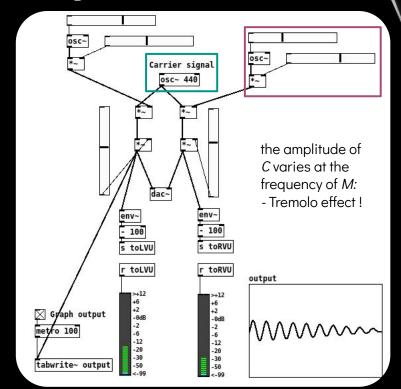
# RM SYNTHESIS Using one oscillator to modulate the gain of an other one

According to your intuition, how would you patch a RM?



tabwrite~ output

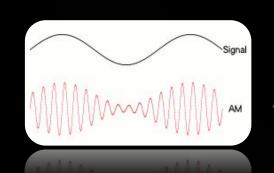
Let's create a tremolo from this: adding RM on both sides



### AMPLITUDE MODULATION

#### Definition:

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).



Earliest method to transmit audio in radio broadcast



Carrier signal high frequency

$$x_p(t) = A_p \cos(\omega_p t)$$



Modulating signal low frequency

$$x_m(t) = A_m \cos(\omega_m t)$$



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

AM

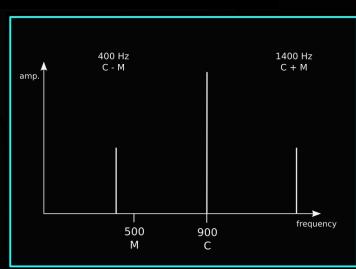
$$A(t) = C(t) \times (M(t) + 1)$$

The modulator M is unipolar, typically set between 0 and 1.

The carrier frequency is preserved and the sidebands generated are at *half* the amplitude of the carrier amplitude.

$$\frac{A_c A_m}{4} \left[ \sin \left( 2\pi \left( f_c - f_m \right) t + \frac{\pi}{2} \right) + \sin \left( 2\pi \left( f_c + f_m \right) t - \frac{\pi}{2} \right) \right] + \frac{A_c}{2} \sin \left( 2\pi f_c t \right)$$

Having the carrier frequency, it is then possible to demodulate the signal in order to access the information hold by the carrier using a pass band.

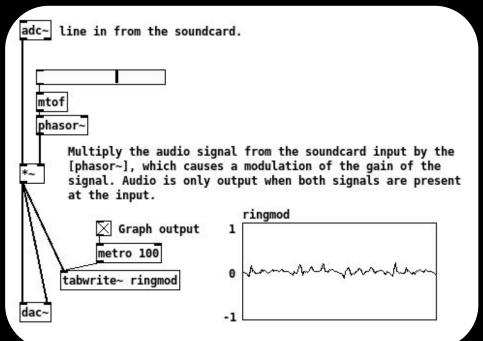


#### AM WITH COMPLEX SIGNAL

Making Alien's voice with ring modulation

Which object will we use to catch our voice?

**→** [adc~]



# FREQUENCY MODULATION

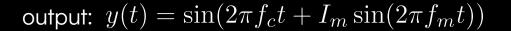


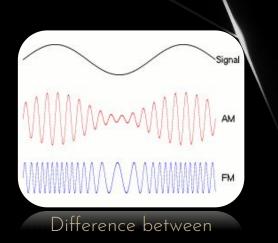
Definition: The information contained in the modulating signal is carried by varying the frequency of the carrier signal.

- 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:





AM and FM

 $f_c$ 

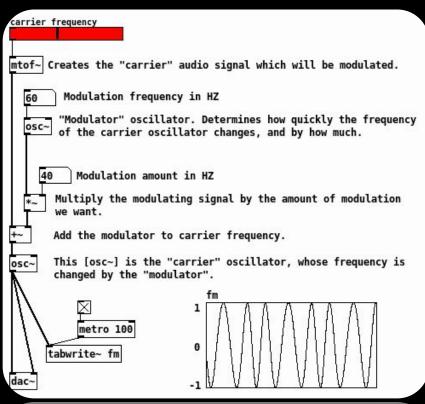
Modulating signal  $f_m$ 

### Frequency Modulation

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

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

#### Simple fm patch



# O3 STEP SEQUENCER

Process Building

#### The Counter

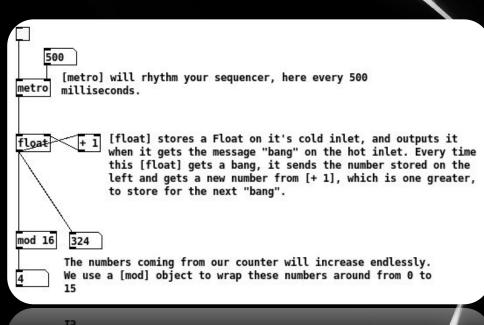
#### Definition of step sequencer:

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 steps sequencer

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



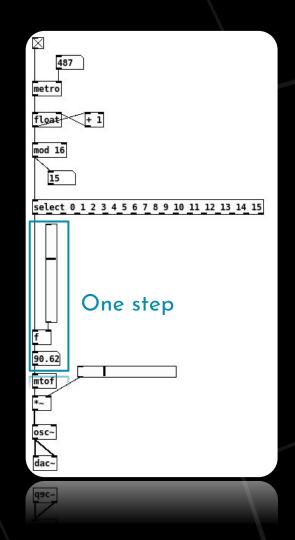
The numbers coming from our counter will increase endlessly. We use a [mod] object to wrap these numbers around from 0 to

# 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...



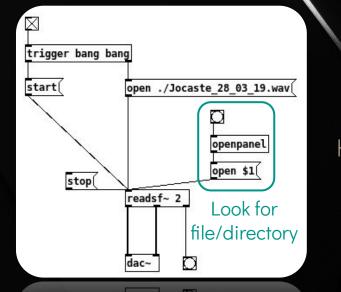
# O4 SOUND FILES

Read & Write

# Read

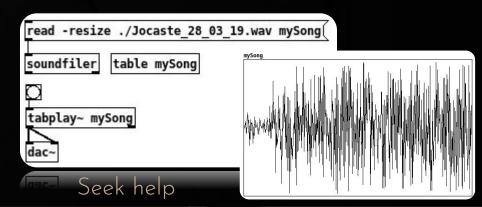
#### Reminder about files formats

- 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





Building of a Kick Drum

# GENERAL APPROACH

Percussive reverb with a white noise

Noise

Resonant Low pass Filter

Volume Env

Square Osc

Pitch Env

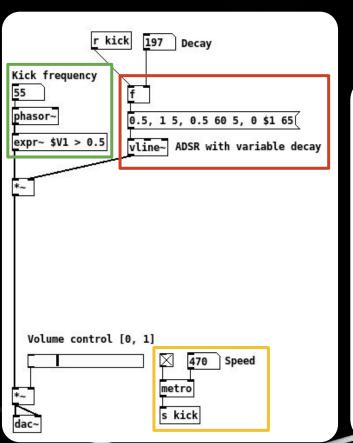
Sine Osc

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

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

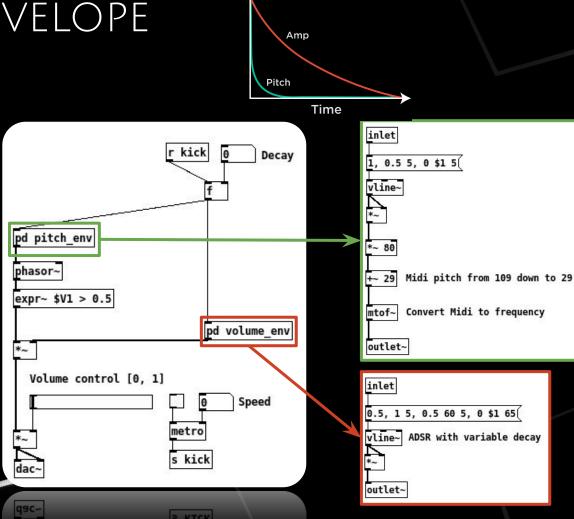
Adding a sine driven by the pitch envelope

# Phasor and Envelope

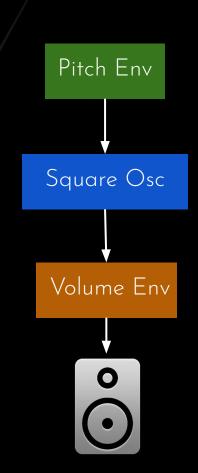


MECK

dac∼



### FIRST STEP ACHIEVEMENT



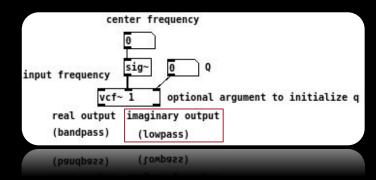
Next:

adding a resonant low-pass filter,

With the cutoff driven by our pitch enveloppe.

Which object do we need?

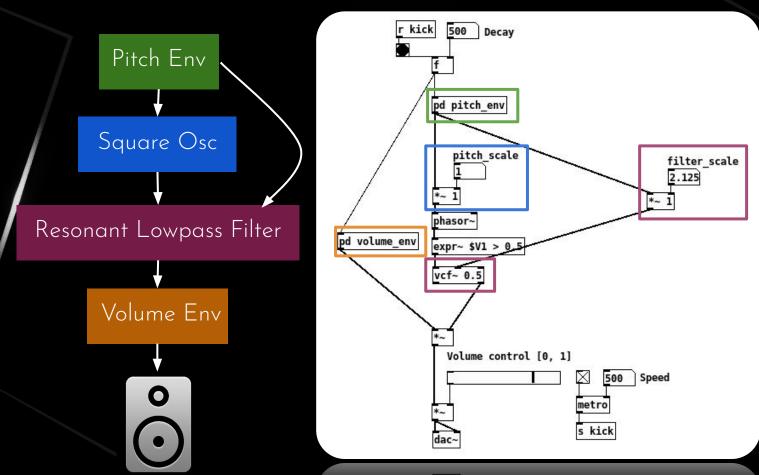
VCF:



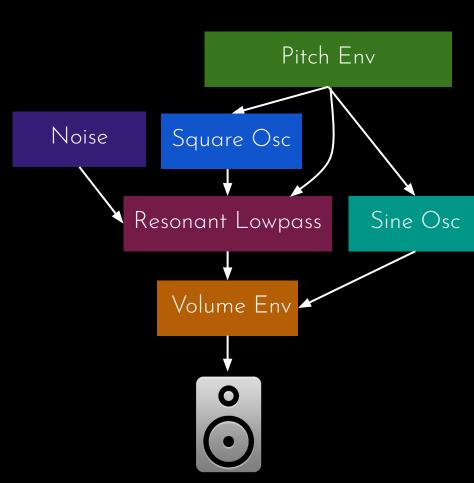
Where will we plug our pitch env?

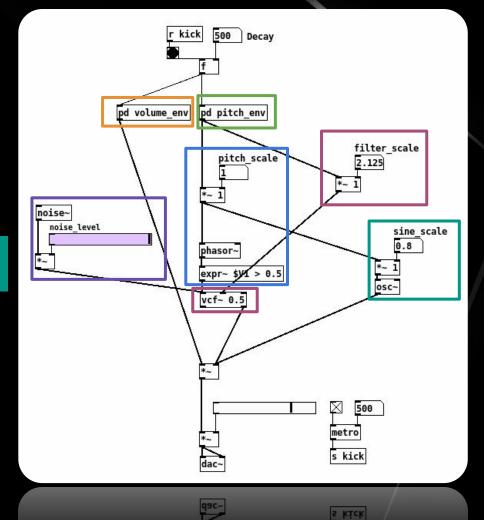
First, let's add a pitch scale, then the vcf with a filter scale.

# RESONANT LOW PASS FILTER & SCALES



# NOISE & SINE

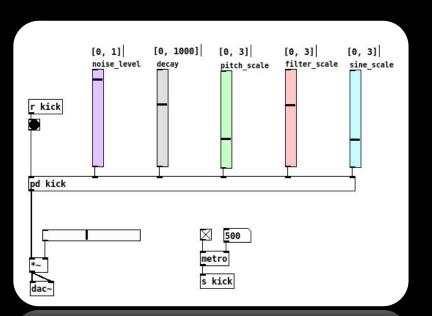


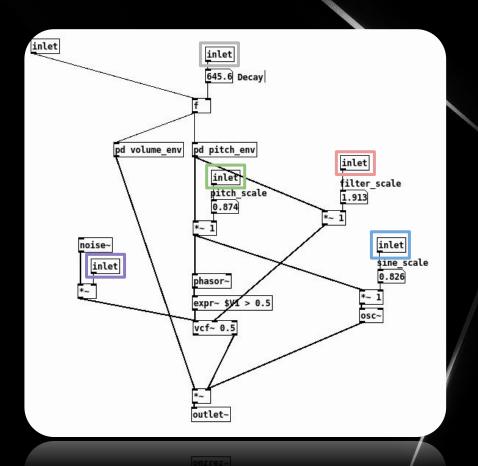


# CLEAN IT UP

#### How to simply subpatch:

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

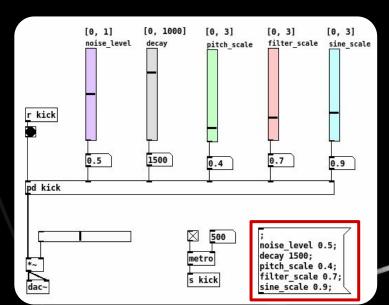


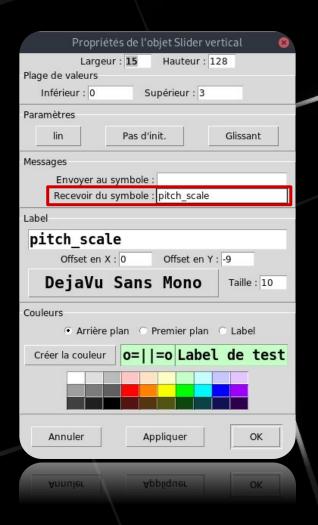


#### ADD PRESETS

#### Messages boxes to add 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



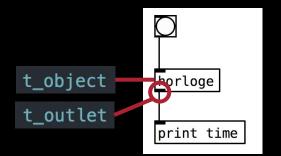


- So what is going on **inside** a given box? So mysterious...
- We can even go deeper (and deeper ... hmm) in Pure Data objects
  - Possibility to define your own boxes :-) Oh woah!
  - The overall system defines PD externals
- PD provides a set of includes and specs
- Simple SDK with a (relatively)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

- 1. 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
# include "m pd.h"
                                               We need to include the PD header definitions
                                               Then define the class of our object
                 *horloge class; -
static t class
typedef struct
                 _horloge
                                           This object reference is mandatory (cf. later)
   t_object
                  x obi:
   t outlet
                   *h out;
                                           Need to manually define inlets and outlets
t horloge;
                                                            3 minimal functions to code
                                                          What happens at runtime (once)
void
                 horloge setup(void); —
                                                         Define object creation (add box)
void
                 *horloge_new(void);
                                                          One method per message
void
                 horloge_bang(t_horloge *x);
                                                    (here we code what happens when a bang is received)
#endif
```

#### Reminder of the data structure

#### 1. Runtime function (\*\_setup)

- Global explanation of the object
- Explains its name, types and functions
- Mimics a class system

Class creation method class\_new

Add the behavior for bang with class\_addbang

Later we will also use class\_addmethod (messages)



#### Reminder of the data structure

#### 2. Box creation function (\*\_new)

- Function called when we create a box
- Similar to an object *constructor*
- Explain all initialization stuff

**Return the created object** 

#### Reminder of the data structure

#### 3. Message handling (\*\_bang)

- Function called when we receive a message
- Here specific example of a bang
- Beware of the processing time!

#### **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)));

}
```

Just get the current time

Need to write symbols to a given symbol table

Write information to a specific outlet

#### What happens for signal stuff?

```
void myfft_tilde_dsp(t_myfft_tilde *x, t_signal **sp)

dsp_add(myfft_tilde_perform, 4, x, sp[0]->s_vec, sp[1]->s_vec, sp[0]->s_n);
}
```

- 1. Our own perform function
- The DSP call (block\_size dependent)
- 3. Memory liberation

Similar class setup method

Need to add the DSP function

DSP call fills the rightful buffers