

# ZYKLØP

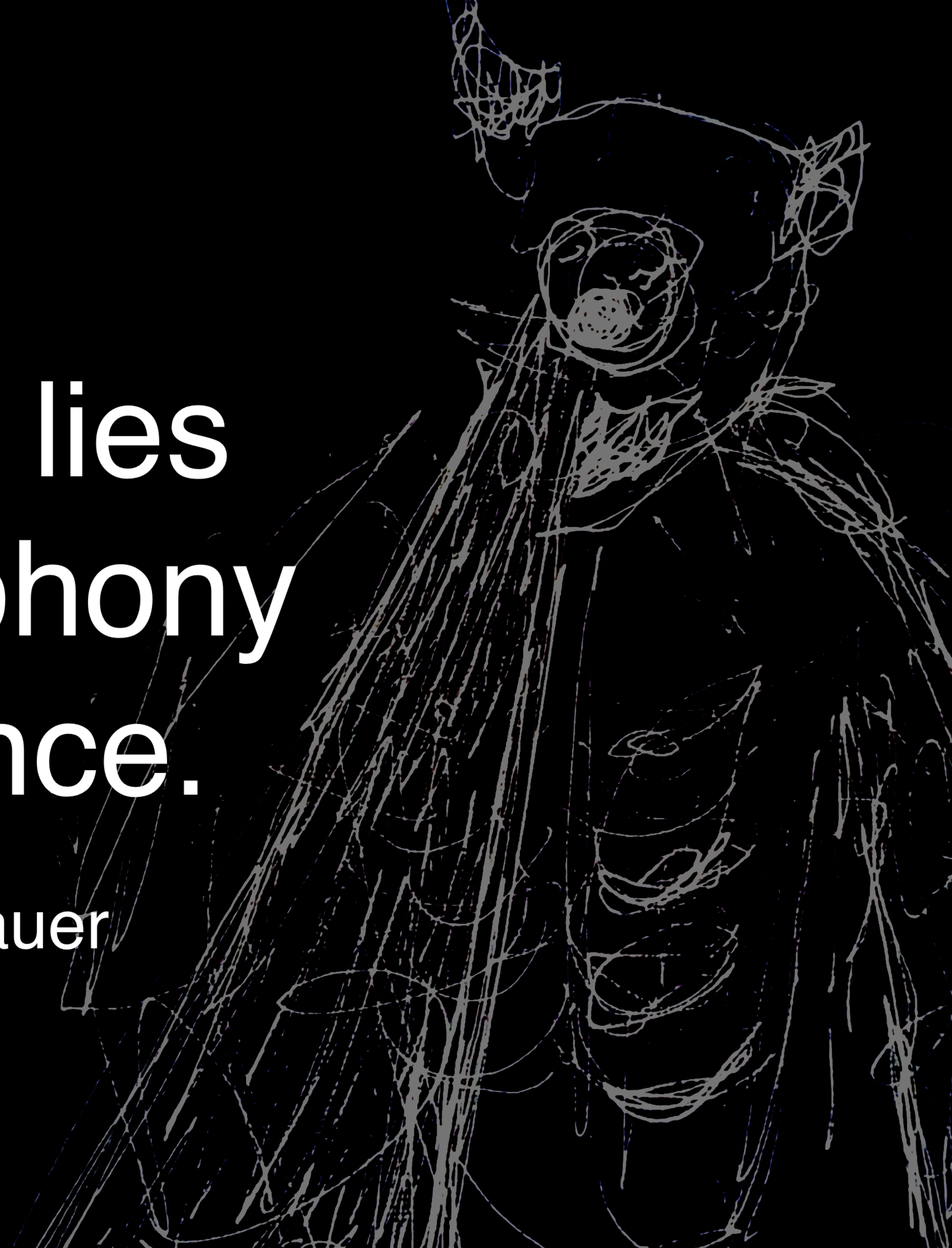


DAWESOME

“

In the Iris lies  
the Symphony  
of Existence.

Arthur Schopenhauer



# THANK YOU



ZYKLØP is a free synth plugin by Dawesome.

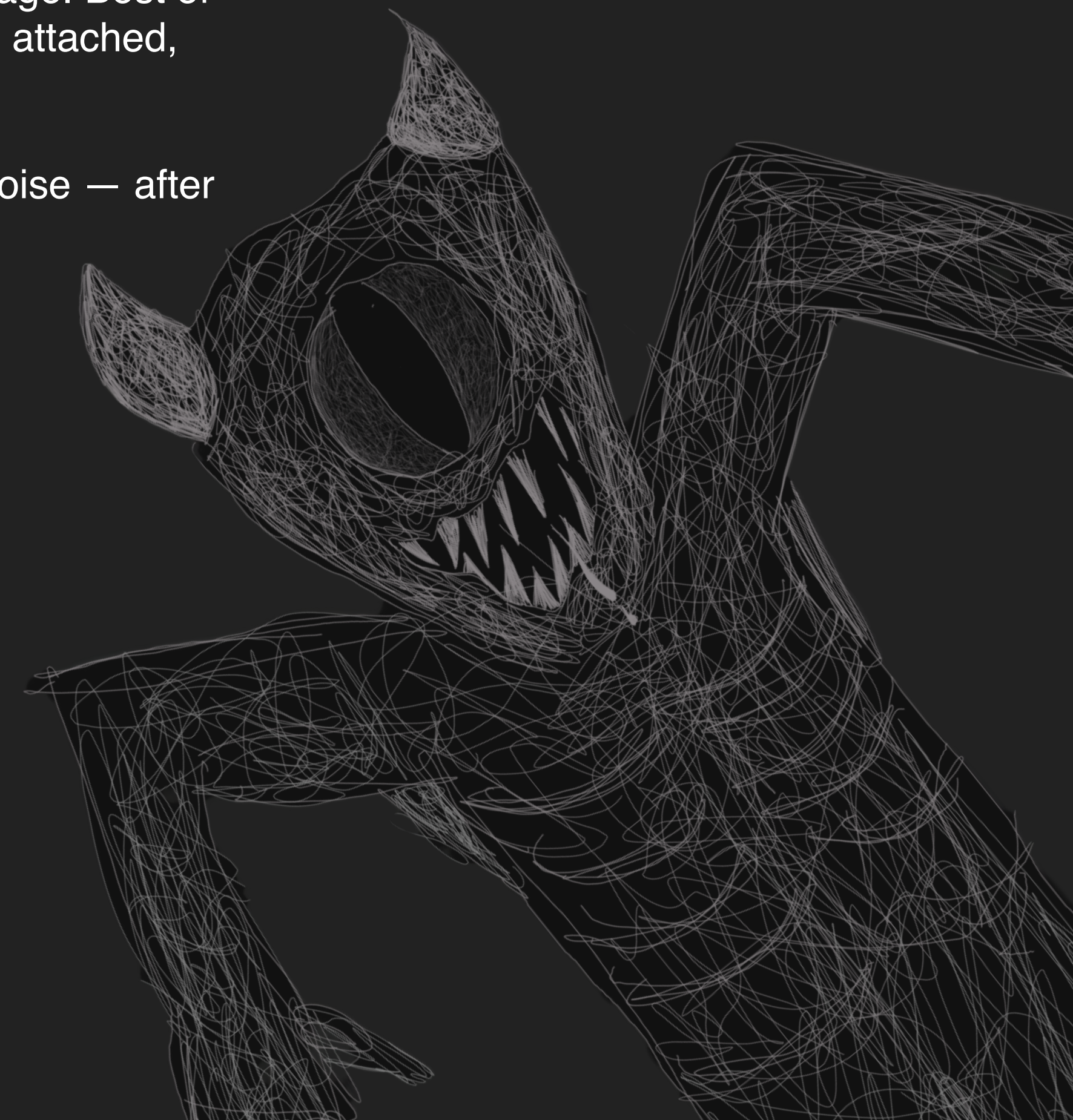
We're all riding the wave of some pretty amazing synth history here... and I figured it's high time to give back to the community that's been awesome to me.

So here's ZYKLØP — the little sibling to MYTH and an absolute blast in a tiny package. Best of all, it's free, which means no strings attached, just pure sonic fun.

Dig in, get weird, and make some noise — after all, isn't that what this is all about?



All the best  
Peter (Dawesome)



# MYTH

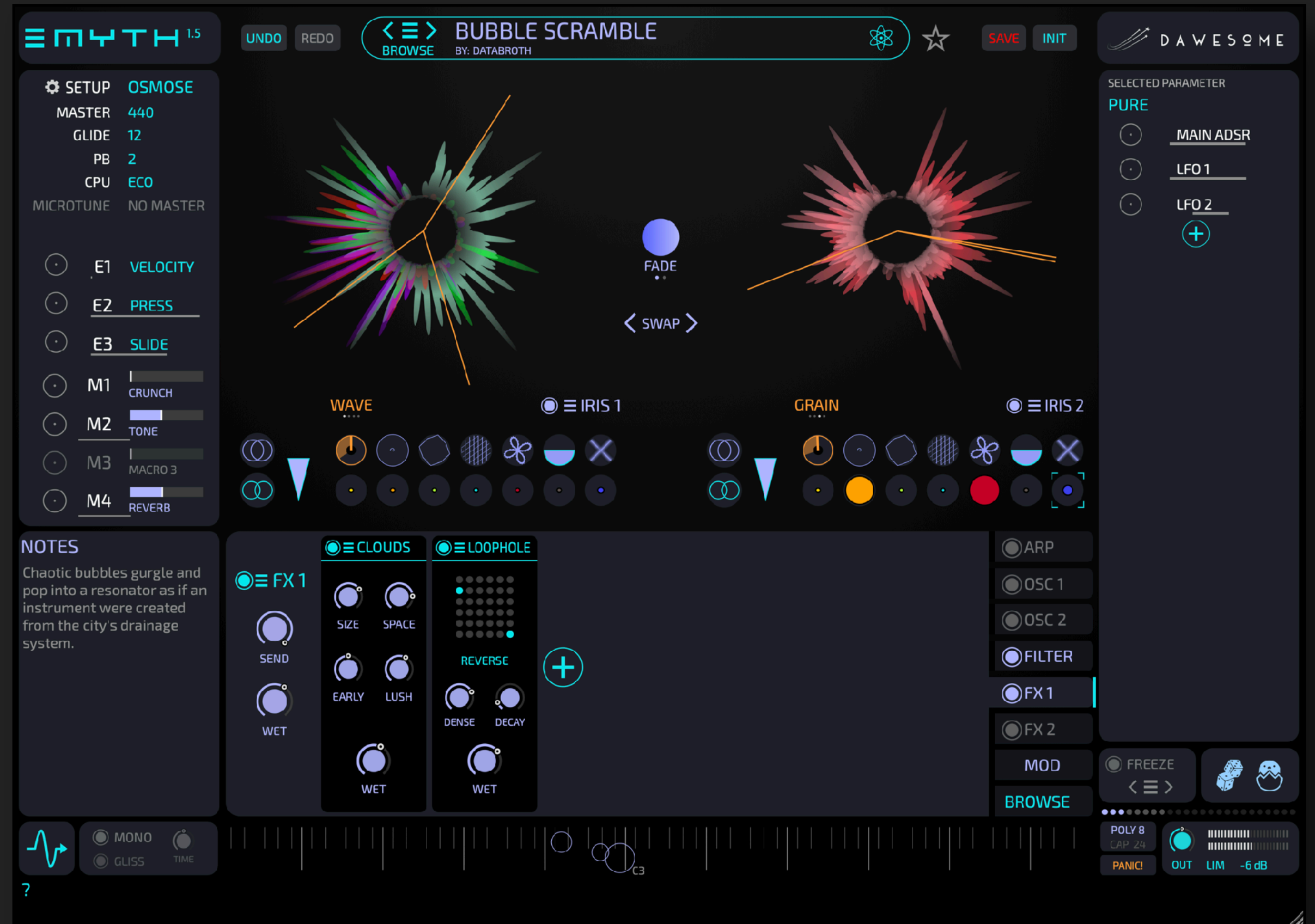
You may have guessed it already ...  
ZYKLØP is MYTH's fun, free-spirited  
little sister.

## Why you might want to go Mythical:

- ▶ Two Eyes, not just one
- ▶ 900+ presets
- ▶ True stereo re-synthesis
- ▶ More modules, 6 per section
- ▶ Two separate FX sections
- ▶ Macros & advanced modulation system
- ▶ 24-voice polyphony

ZYKLØP's a blast ...

but if you're ready to go big, "Myth" is where it's at!



[MYTH is available via tracktion.com](https://tracktion.com)

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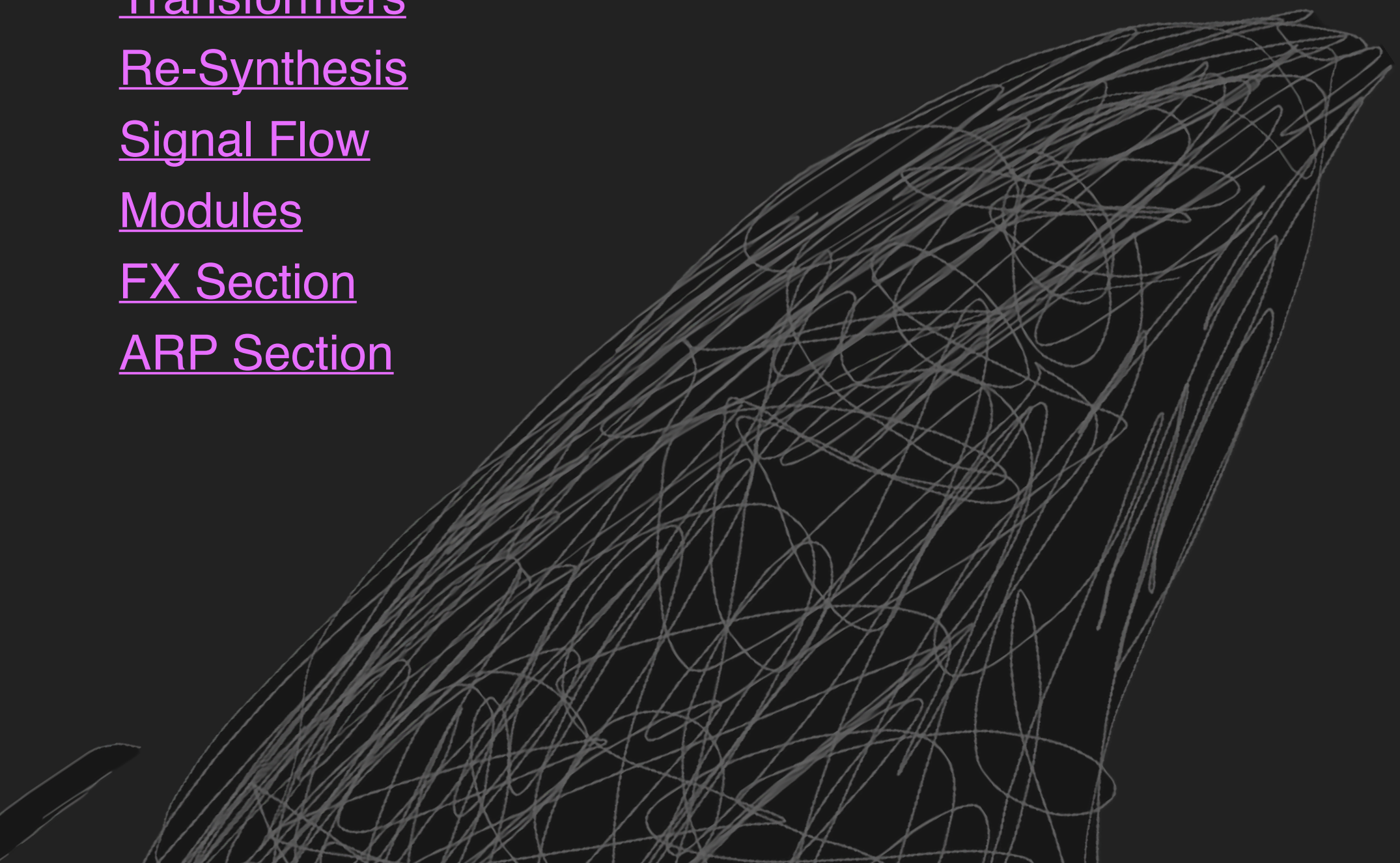
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# GETTING STARTED

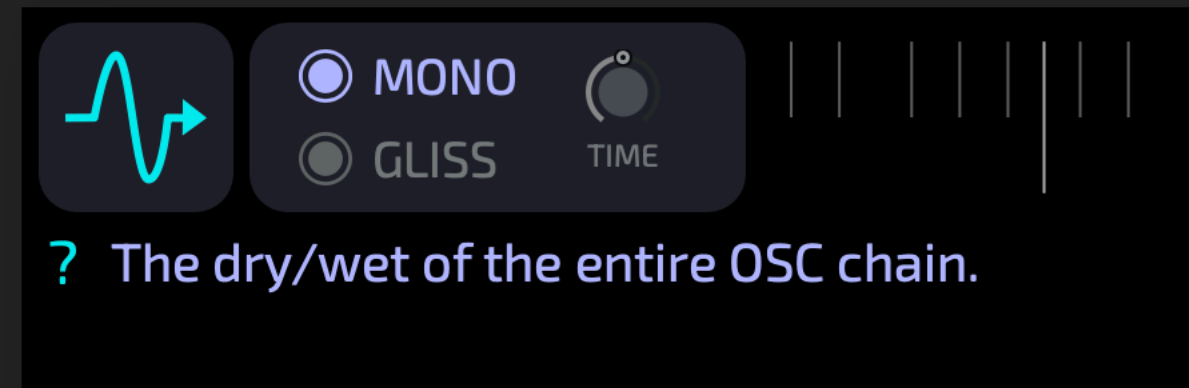
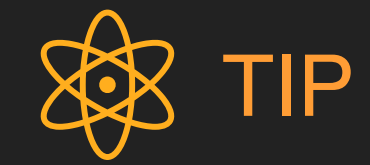
This involves two steps:

## 1 Install the software

This is straight forward: download the right installer for your system:  
For example: [Zyklop\\_1.00.pkg](#) for Mac or [Zyklop\\_1.00.exe](#) for Win.  
You can start the installer with double-click.

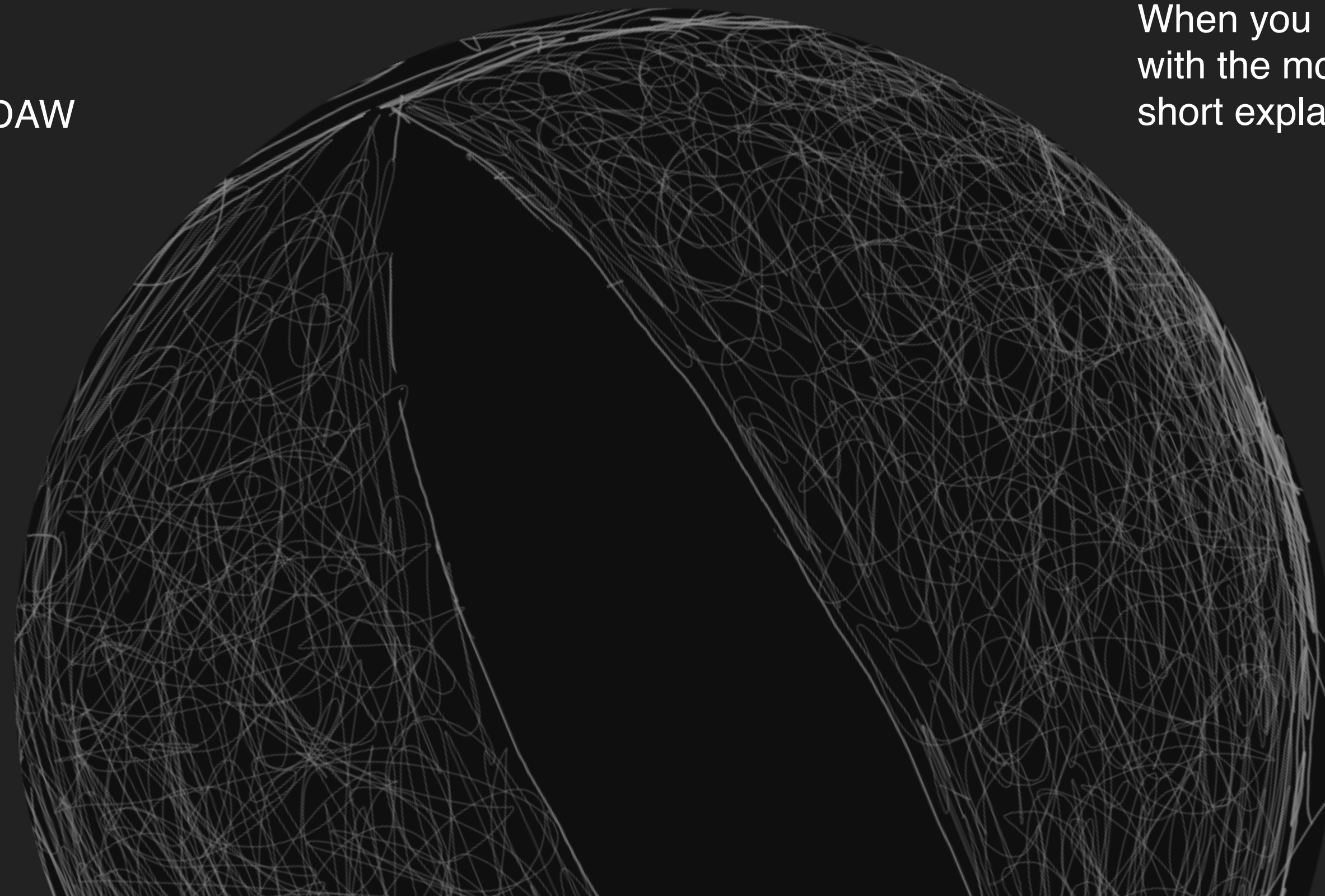
## 2 Have fun!

Open as many instances of **ZYKLØP** in your DAW  
as your mom will let you



**ZYKLØP** has built-in help. You can de-/activate it with the question mark **?** in the lower left corner.

When you hover over any element with the mouse you will see a short explanation.



# THE SETUP

The word **SETUP** may not sound very sexy, but this is really important.

Nowadays there are a variety of midi-controllers to choose from, and these come with different methods for expression, e.g., velocity, mod-wheel, MPE, aftertouch.

This creates a problem: the sound designer who created presets most likely used a different controller than you.

ZYKLØP addresses this via the **SETUP**. You just need to tell ZYKLØP what kind of expressive midi data is sent by your controller.



Important point: this information is stored for the plugin, not with the preset. Hence you need to choose your **SETUP** only once, and then you can browse presets and they will always match to your controller.

Click the gear symbol to choose a pre-defined setup.

You can also save the current setup.

⚙️	SETUP	SEABOARD
	MASTER	440
	GLIDE	12
	PB	2
	CPU	ECO
⦿	E1	VELOCITY
⦿	E2	PRESS
⦿	E3	SLIDE

You can also configure freely to your specific needs. Right-click to choose the midi input for the main *expression-inputs*

You can then give this configuration a name and use the **SETUP** menu to store it.



## TIP

There are two different pre-defined setups for the OSMOSE. The first one has E2 = PRESS and E3 = SLIDE. The second one uses an input source called OSMOSE which combines PRESS and SLIDE to one gesture.

# THE SETUP - for Sound Designers

Acoustic music instruments do not simply switch notes on and off - the performer usually plays notes with *expression*.

How this is performed and how this sounds is different from instrument to instrument. On a piano you vary the speed and force when hitting a key, on a wind instrument you control the speed and pressure of the air flow and the size of the oral cavity.

ZYKLØP uses three specific modulation sources for expressive playing:

E1 - for *NOTE-ONSET* expression

E2 - for expression via *INTENSITY*

E3 - for expressing with *TIMBRE*

How the musician physically *performs the expression* depends on their midi controller - hence this is configured here in the **SETUP**. This is the same for all presets.

How the preset *reacts to the expressive midi data* and how the sound is affected by it are saved as part of the preset.

## TIP

1. For E1 you should always assume **Midi-Velocity**
2. Increasing E2 should increase the perceived intensity of the sound - in most controllers increasing E2 means that you apply more force. It feels very unnatural if applying more physical force leads to a softer tone.
3. On the OSMOSE the SLIDE is activated by pressing the key further down. That means that E3 increases only once E2 reached its maximum. Hence E2 should not create complete tonal mess if you want to offer E3

Some examples of typical controller setups that you can find in the wild: The **SETUP** mechanism of ZYKLØP allows you to make sounds that translates from your setup to the users setup.

	Midi	MPE	OSMOSE	User A	User B	...
E1	Vel	Vel	Vel	ModW	Vel	
E2	ModW	Press	Press	Pedal 1	ModW	...
E3	Pedal	Slide	ModW	Pedal 2	PitchBend	

# OVERVIEW

The image shows a screenshot of the ZYKLØP software interface, which is a digital synthesizer. The interface is dark-themed with various controls and sections. Annotations with white lines pointing to specific elements provide an overview of the software's features and controls.

**Annotations:**

- UNDO and REDO of last operation**: Points to the UNDO and REDO buttons at the top left.
- Go to next or previous patch of your current selection. [More info here.](#)**: Points to the BROWSE button at the top center.
- Click to save the patch. Turns red if there are unsaved changes. [More info here.](#)**: Points to the SAVE button at the top right.
- Choose a random preset.**: Points to the DAWESOME button at the top right.
- Randomise the preset. Press longer for more changes.**: Points to the cube icon in the preset list.
- The IRIS is a core piece of the synthesis engine. You can drag and drop samples here to re-synthesize them. [as explained here.](#)**: Points to the central IRIS visualization.
- TRANSFORMERS are used to shape the sound [as explained here.](#)**: Points to the WAVE section at the bottom center.
- The UI is resizable. Drag to shrink or expand**: Points to the bottom right corner of the interface.
- The ZYKLØP engine has multiple SECTIONS - [explained here.](#)**: Points to the J-60, CLOUDS, and LOOPHOLE sections at the bottom.
- ZYKLØP offers various MODULES, [these are explained here.](#)**: Points to the FX, SEND, and WET controls at the bottom.
- The virtual keyboard displays incoming midi notes**: Points to the keyboard at the bottom.
- Click to enable or disable tooltips**: Points to the question mark icon at the bottom left.
- You can drag and drop the last played notes as audio from here.**: Points to the waveform icon at the bottom left.
- You can enter notes for your preset.**: Points to the NOTES section on the left.
- The modulation sources used in the preset. [Explained here.](#)**: Points to the modulation sources on the left.
- It's explained in detail [here.](#)**: Points to the STRING section on the left.
- To get the most out of ZYKLØP you need to tell her what kind of midi setup you are using.**: Points to the MIDI setup section on the left.
- Click the cogwheel to change the SETUP**: Points to the cogwheel icon on the left.
- Main menu** : Points to the main menu icon on the left.

# BROWSE & PLAY

The easiest way to browse presets is to use the < and > buttons. This always refers to the current selection of presets you made on the browse tab.

Click **BROWSE** to open the browse tab - this can be used to filter for a specific subset of presets as explained on the next page.

In the **BROWSE** tab there are up to four filters. The first two are for PACKS and their subdirectories.

Afterwards you can add up to two more filters.

As displaced here, only presets from an AROVANE folder are selected.

Only presets that are tagged as COLD pass this filter.

The burger menu ☰ offers quick access to a specific preset.

Click the star symbol to add the current preset to your favorites.

PACK	SUBDIR	TYPE	TONE
USER AROVANE	<NO SUBDIRS>	ATMO BASS DRONE FX KEYS MALLETS PAD SEQUENCE STRING SYNTH TEXTURE	BRIGHT CLEAN DIRTY THIN PHAT WARM COLD SOFT HARD LIGHT AIRY

Click to switch to a preset.

Click on the star symbol to add a preset to your favorites.

Right-click for a few more options.

LOAD

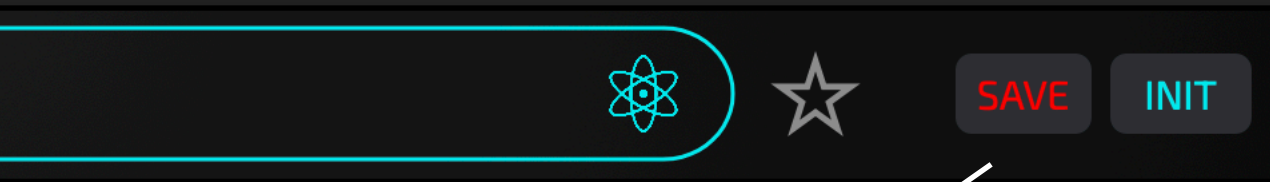
REVEAL IN DIR

MAKE THIS THE INIT PATCH

You can specify the sorting order.

You can also filter with full text. Below you can set which text areas should be looked at.

# SAVE PRESETS



The **SAVE** button indicates with **red** colour if there are unsaved changes.

When you click **SAVE** a special window opens.

Review and/or edit your preset notes.

Close the window without saving the preset. The tags and preset notes are stored internally and are not discarded.

Click to change the name of the preset

Click to choose a different subdirectory for saving

### SAVE PRESET

PRESET NAME: **FACE HUGGER**

AUTHOR: **DAWESOME**

DIRECTORY: **≡ USER PRESETS/**

NOTES:  
ENTER TEXT

TYPE	CHARACTER	TIMBRE	TONE	MOOD	MOTION	SHAPE
ATMO	EPIC	METALLIC	BRIGHT	JOYFUL	STATIC	PERCUSSIVE
BASS	CINEMATIC	GLASSY	DARK	LIVELY	ORGANIC	SLOW ATTACK
BRASS	ENIGMATIC	NOISY	CLEAN	MYSTERIOUS	SIMPLE	FAST ATTACK
DRONE	SYNTHETIC	DULL	DIRTY	EERIE	COMPLEX	PLUCKED
FX	MYSTICAL	RESONANT	THIN	SCARY	PULSING	SUSTAINED
KEYS	FUTURISTIC	HARMONIUS	PHAT	TRANQUIL	EVOLVING	SHORT
LEAD	VINTAGE	INHARMONIC	WARM	TENDER	HYPNOTIC	LONG TAIL
MALLETS	MODERN	DISSONANT	COLD	RELAXING	STUTTER	LEGATO
PAD	DRAMATIC	RASPY	SOFT	NEUTRAL	GLITCHY	SLURRED
SEQUENCE	AMBIENT	GRAINY	HARD	NERVOUS	ECHOING	STACCATO
STRING	URBAN	MELLOW	HEAVY		CHAOTIC	ARPEGGIATED
SYNTH	WTF	DEEP LOW	LIGHT		SLOW MOTION	
TEXTURE			AIRY		FAST MOTION	

**CLOSE** **OVERWRITE** **VERSIONIZE**


This button appears if a preset with the same name already exists.

When you hit **VERSIONIZE** a number will be added to the preset name to avoid overwriting the existing preset.



**TIP**  
Via the **≡** symbol, you can reveal the directory where the user presets are stored. Use this to create new subfolders to organize your presets.



**TIP**  
Sometimes it's hard to come up with a good name. Click on the  symbol to invent a name.

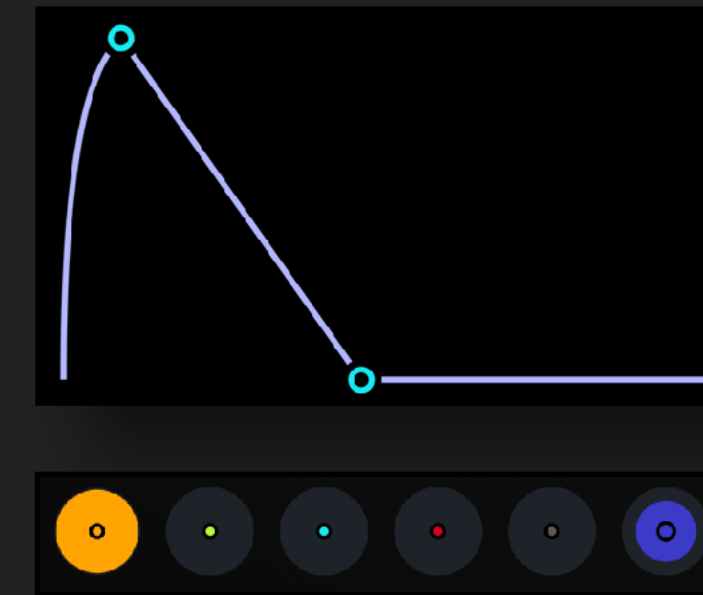
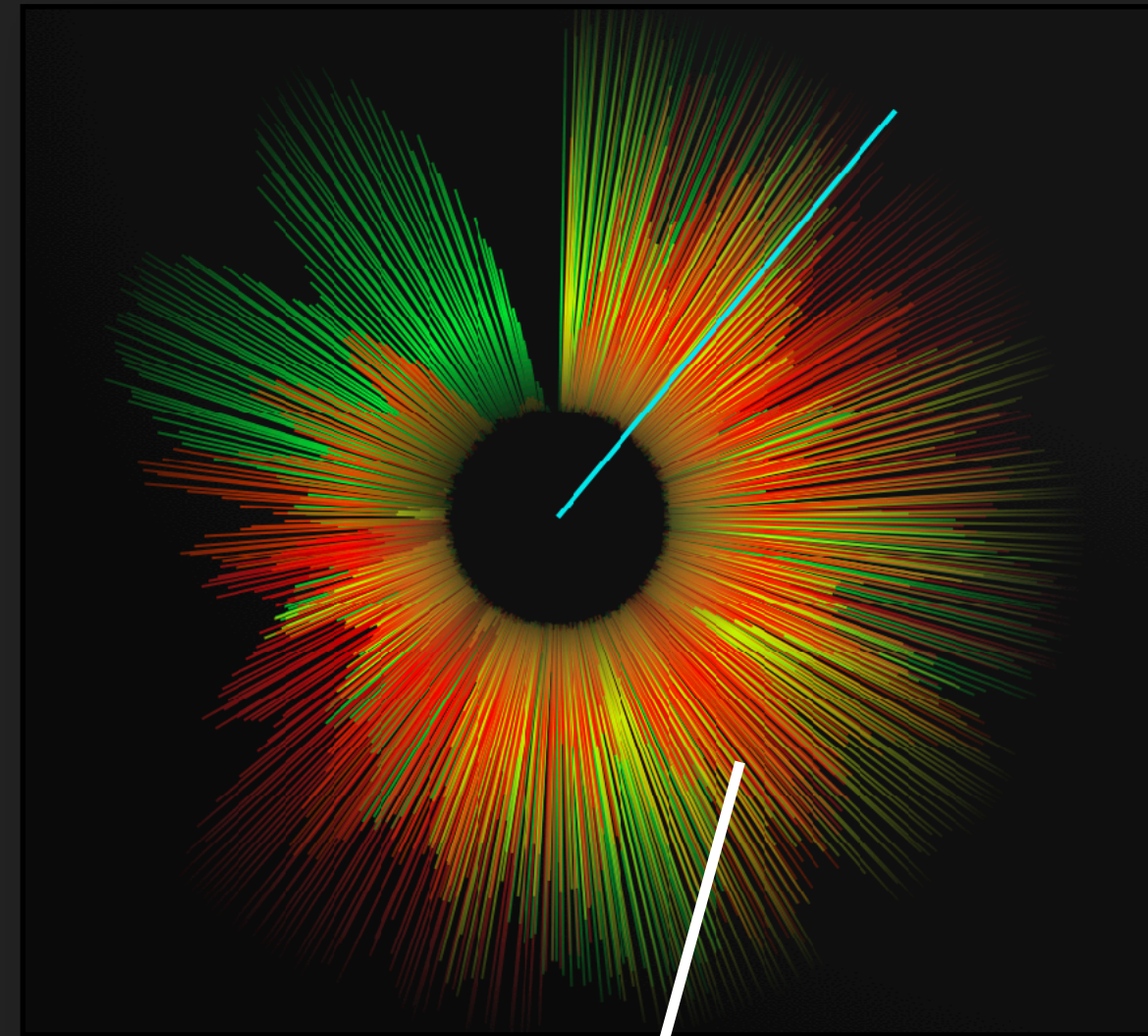
Right-click will provide you with a list of ten different name options. Often these names are not perfect, but spark a good idea.

# SOUND DESIGN - INTRO

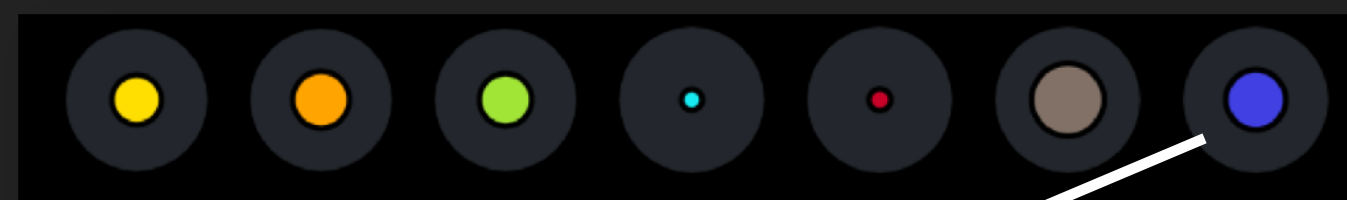
The IRIS is *not a sampler* - instead you should better think of it as an oscillator derived from an audio source.

Oscillators need to have a lot of overtones that can be filtered afterwards with a filter in subtractive synthesis, or to be used as an exciter for physical modelling.

Hence you will often hear, that the IRIS first sounds very harsh and noisy. This is by design ... and you will see: this is your friend!



Modulate the **BLUE** transformer with a smacky ADSR such that you have no BLUE at attack and during decay it adds more and more blue. In this way you can shape natural sounding transients.



Use the **BLUE** transformer to make the IRIS softer. Then you can use the other transformers to add new overtones. Why is this useful? You can now modulate the various colours of your sound!



Hold a key on your controller and now you can click and drag on the **IRIS** to find a sweet spot. Or modulate the **POS** for example with an **ADSR** or a **LFO**



You can also keep the "harsh" sound (or even make it stronger with the transformers) and then use its power to drive a **MODAL** or **RESO** module. You can drag and drop your own samples to use as timbres - an easy route to physical modeling.



# THE IRIS

In classical synthesis, oscillator waves are mostly static and you add filters and modulators to “program” variations that make the sound interesting.

ZYKLØP offers a paradigm shift: it uses machine learning to analyse your sound material and approximate it with a *synthesis engine* displayed as an IRIS.

The inherent variations and richness of the sample and its temporal changes can now be exploited and transformed.

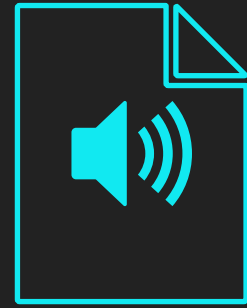
Instead of “programming” sounds you create sounds by exploration and tweaking knobs. Of course - you can also add classical LFOs, envelopes or filters as much as you want. Just ... you will rarely feel the need!



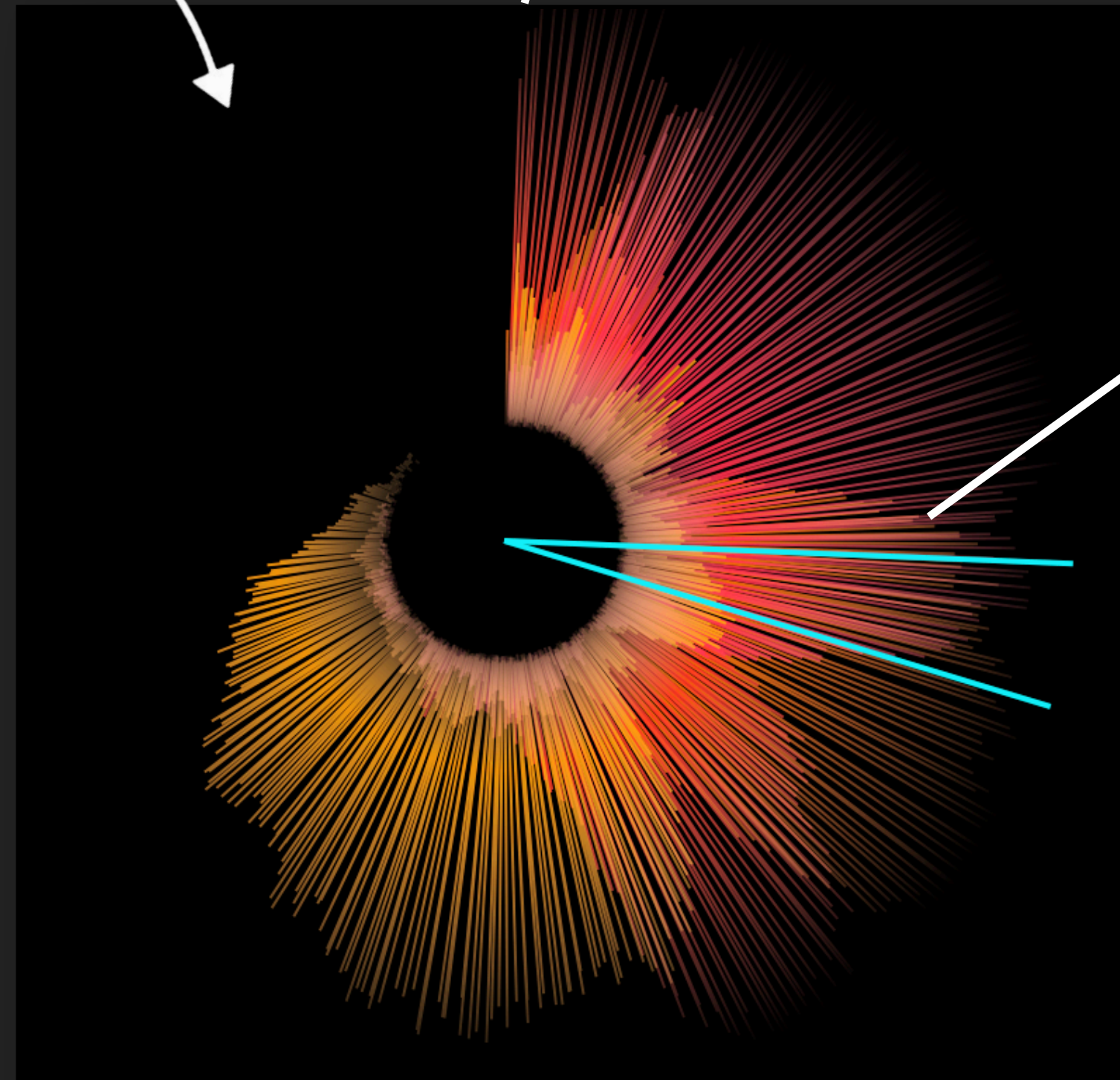
TIP

The easiest way to tweak an existing preset: just drop your own samples onto the IRIS!

Drag and drop your own samples onto the IRIS



The IRIS is a representation of a sound source as a loop.  
The start and end of the sample are at the 12 o'clock position.



The play position for each note is indicated with a turquoise line.

You can click-and-drag the IRIS while playing to change the position.



TIP

Often, you'll want to modulate the play position with an LFO. You can right-click the IRIS and choose SET AND USE LFO from the context menu. This will set the RATE of the LFO to match the duration of the sample and create a modulation of the POS by the LFO.

# TRANSFORMERS

Transformers work as simple as the cutoff dial of a filter - they modify the sound and can be modulated.

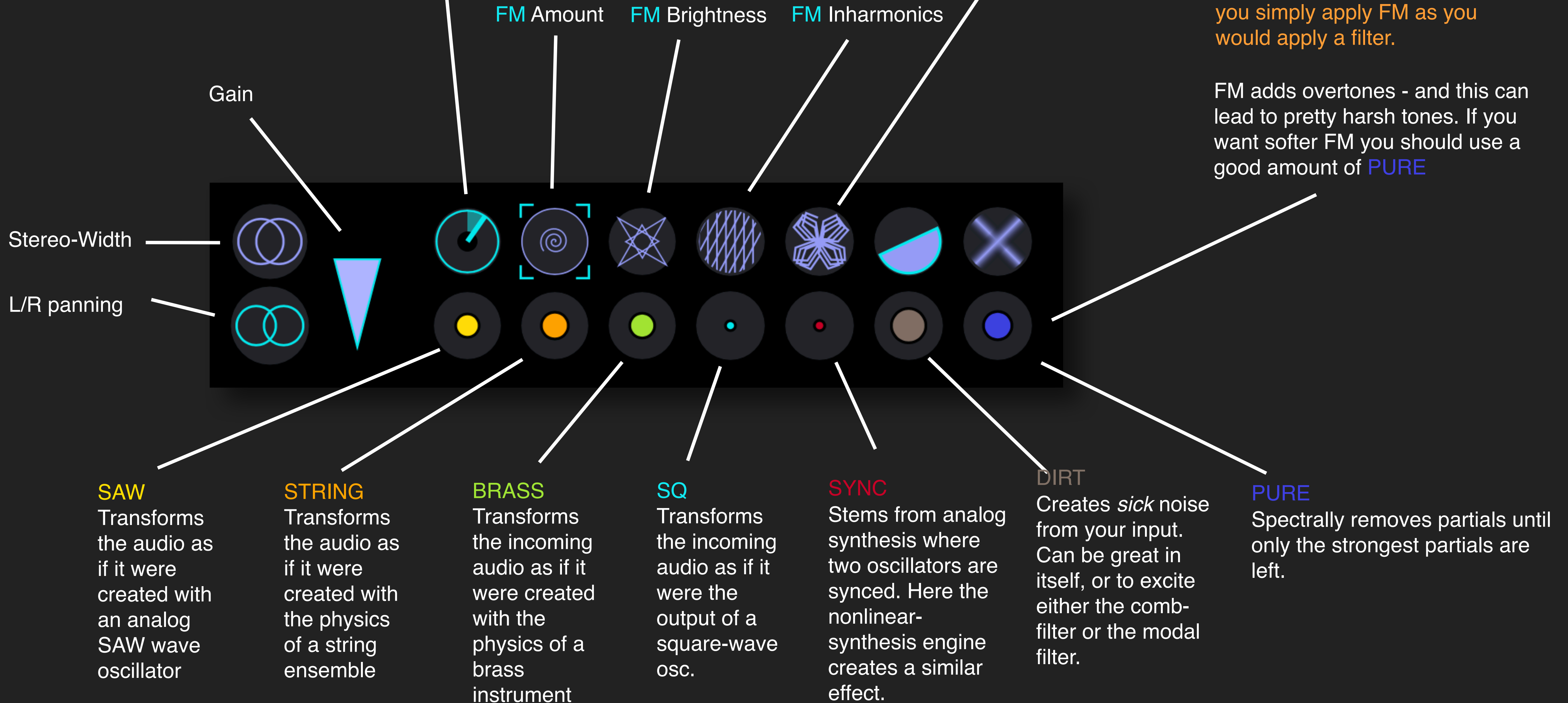
The position of the "playhead" on the IRIS. The value range is two complete turns.

Animates the sound. Large values detune the sound.



FM synthesis sounds great - but "algorithms" and "operators" are not your cup of tea? In ZYKLØP you simply apply FM as you would apply a filter.

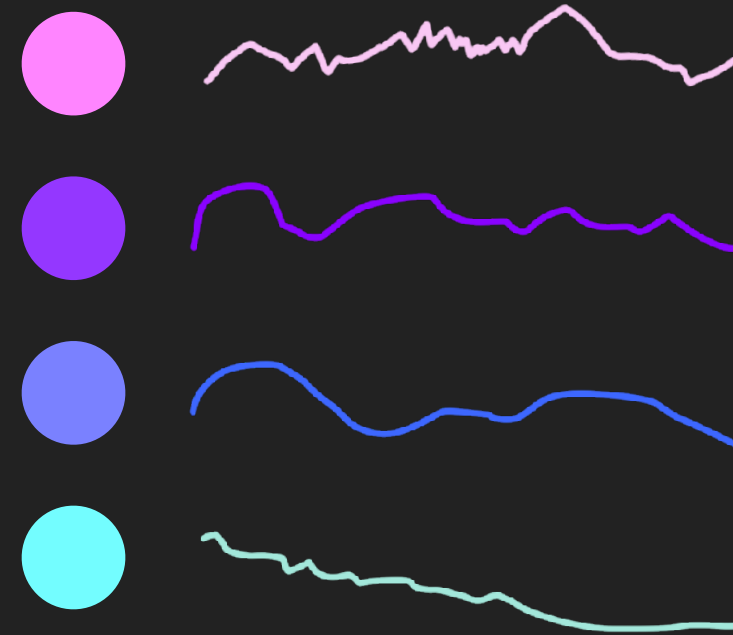
FM adds overtones - and this can lead to pretty harsh tones. If you want softer FM you should use a good amount of PURE



# RE-SYNTHESIS

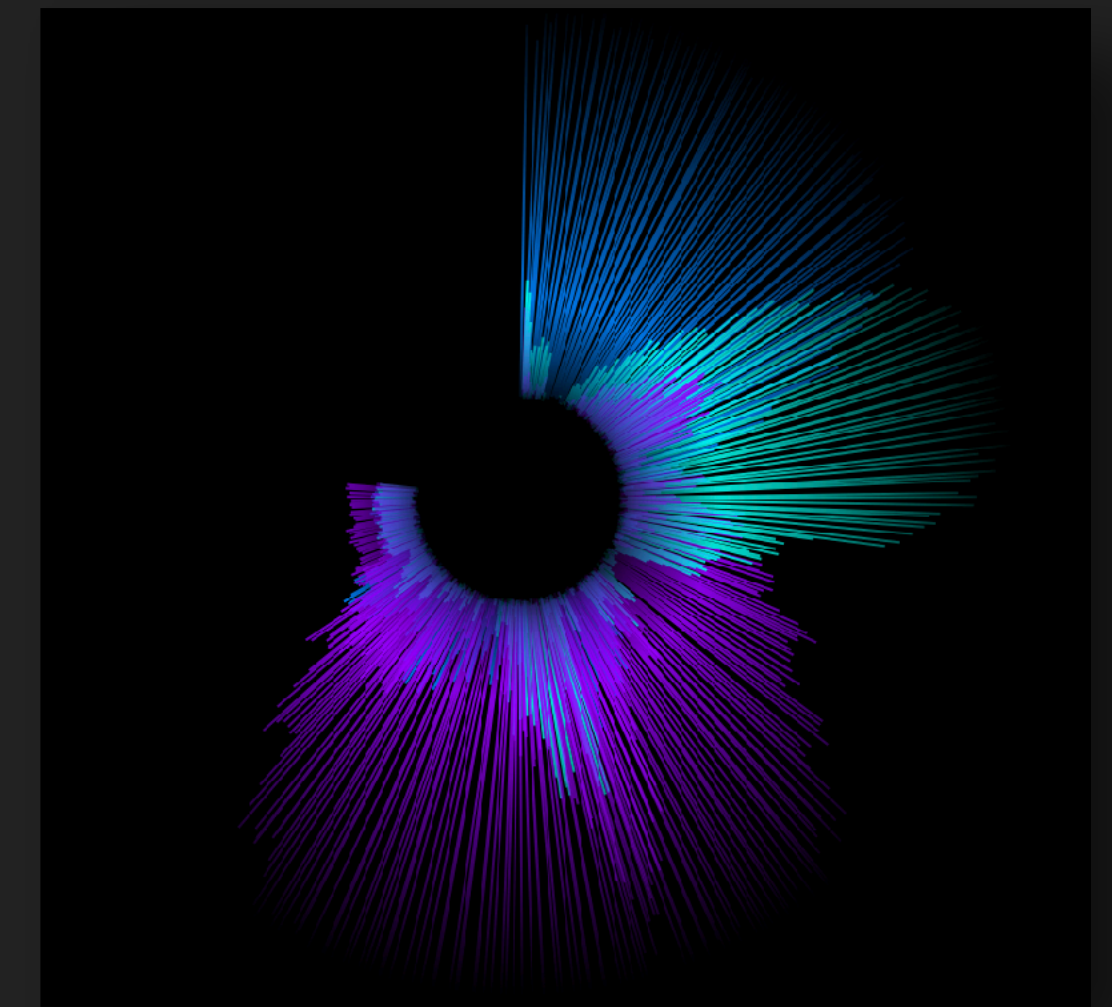
When you drop audio onto the IRIS, machine learning is used to re-synthesize the sample in a special way.

ZYKLØP approximates your sample by finding the optimal combination of four static sounds whose volume is controlled by an envelope curve.



Each of the four internal oscillators of the IRIS is capable of creating any static sound.

Each of the envelopes is a MSEG with up to 500 segments.



Together these are represented as an IRIS

There are two more transformers.

These don't act on the audio stream like the other transformers, instead they change the temporal evolution.

## WTF

WTF changes the volume of all four internal oscillators at the same time.

If you turn it left, the brightest oscillators are faded out. If you turn it right, the darkest oscillators are.



TIP

Like all the other transformers - everything can be modulated and automated.



## SMOOTH

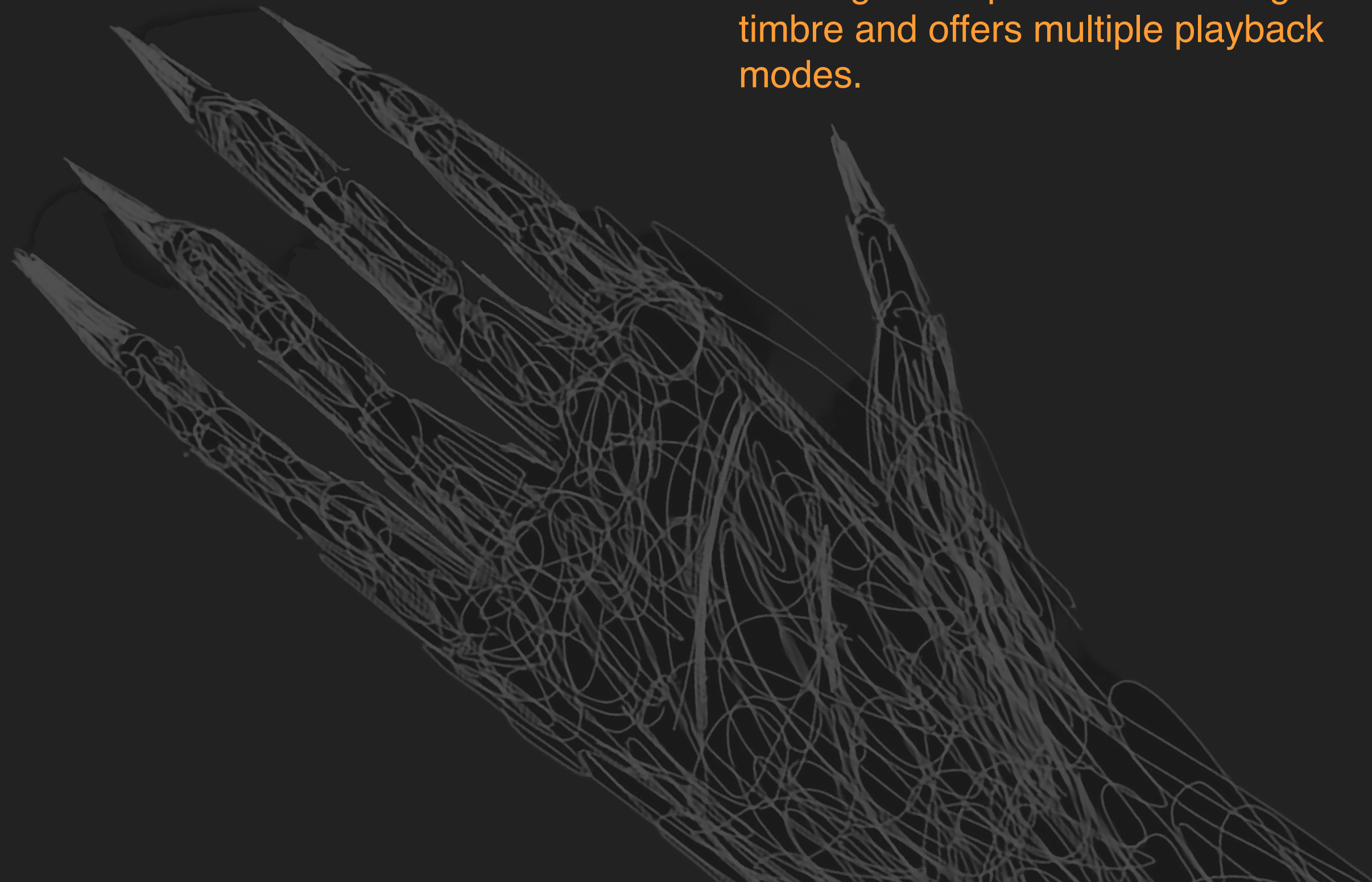
This simplifies the envelope curve, making it smoother.

# RE-SYNTHESIS ALGOS

You import audio by drag and drop onto the IRIS.

The original re-synthesis algorithm V1 was optimised to extract timbral changes, but it is not very faithful to the original timbre in many cases.

The V2 re-synthesis algorithm, optimised to preserve the characteristics of the source material.



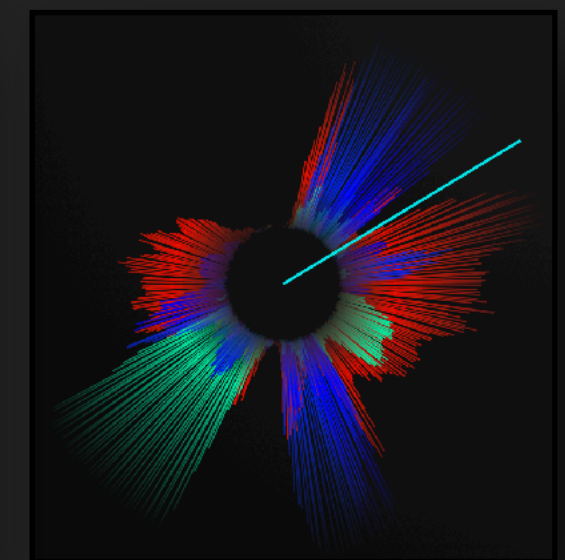
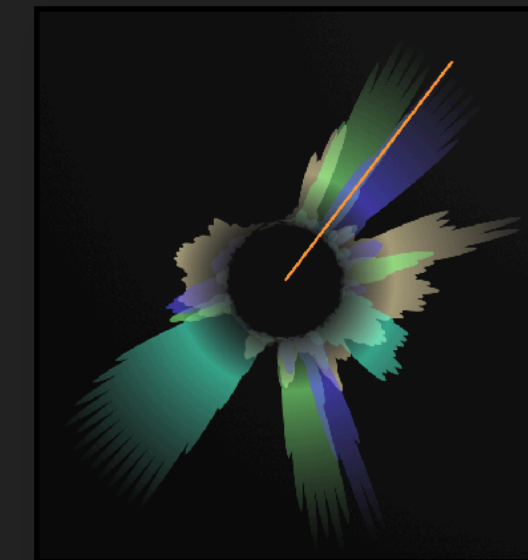
Drop audio here for the new re-synthesis algorithm.

This algorithm preserves the original timbre and offers multiple playback modes.



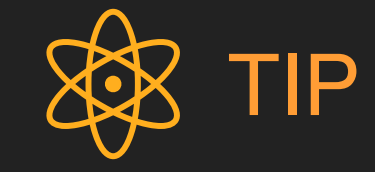
Drop audio here for the classic re-synthesis algorithm.

The appearance of an IRIS and its playhead will differ depending on which re-synthesis style was used.



# PLAYBACK MODES

If your IRIS was imported with the new re-synthesis algorithm you can choose between four playback modes.

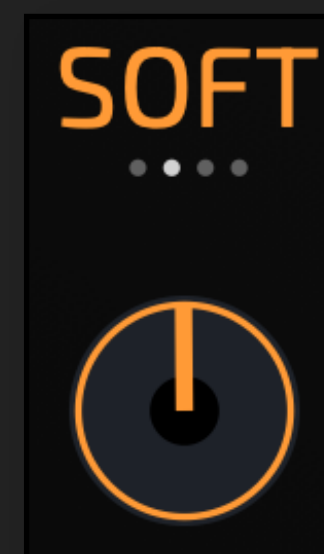


Changing the playback mode is fast and simple, and sometimes a different playback mode gives you pleasant surprises.



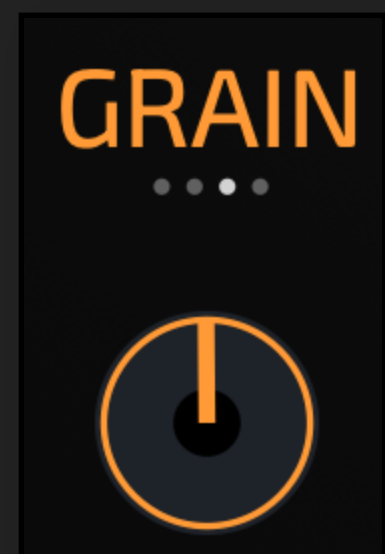
Works great on wavetables and typical synth sounds.

You can export wavetables from any wavetable synth with note F, -1 octave and -23 cents. These settings usually give good results in ZYKLØP.

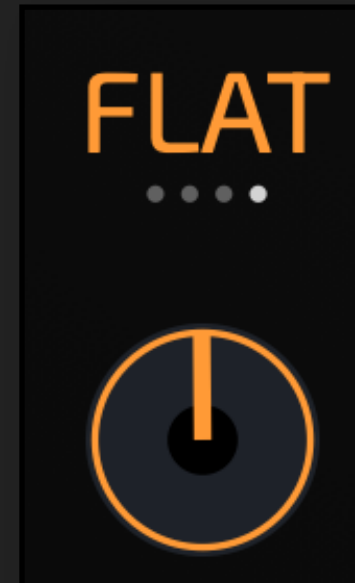


This is the "general purpose" mode for acoustic instruments or voice.

Also gives good results at very slow speeds.



Gives very interesting result on some sources, for example drum loops.



This mode tries to flatten out pitch changes.

If the signal does not contain a clear pitch, is inharmonic, noisy or the weather conditions are bad the results may not be what you expect.

# SIGNAL FLOW

IRIS

OSC

FILTER

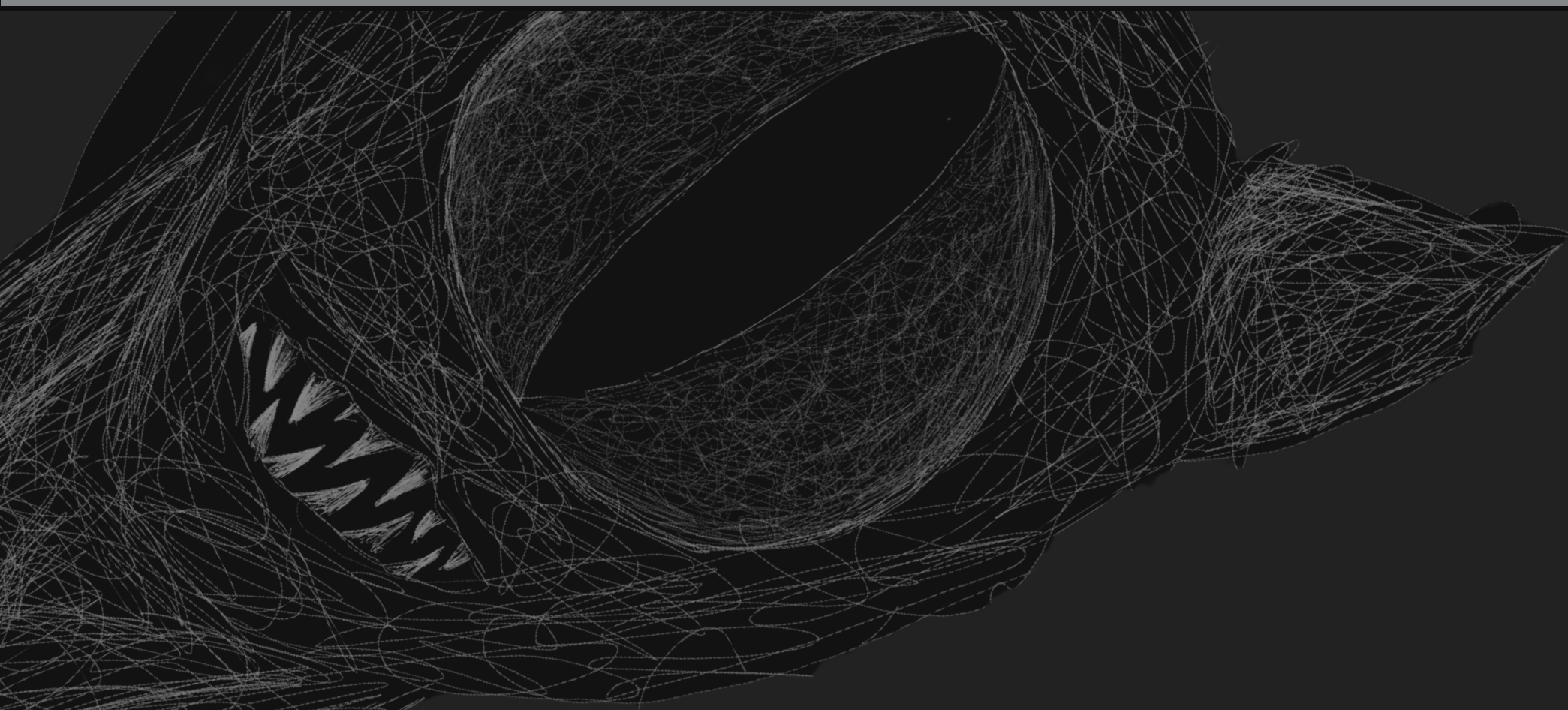
FX

OUT



The IRIS, OSC and FILTER sections are per voice

The voices are summed and feed into the FX



# MODULES

Modular synthesis can be very flexible, but also can get very complex with all the cables.

ZYKLØP offers modules for great flexibility and sound shaping. The system takes inspiration from classical pedal boards - where the effects are simply *chained*.

The menu ≡ symbol offers multiple functions. For example, you can reset a module to its defaults or save the state as a named "sub-preset" for this module

Drag and drop a module by its name label to re-order. You can also revoke a module by dragging it out of view.

Click on the + symbol to add another module. Each section can have up to 3 modules.

Enable / disable the entire chain by clicking on ●

The menu ≡ symbol offers multiple functions. For example, you can remove all modules with one click, or you can save / restore the entire section as a "sub-preset".



Click on any tab to switch to the currently displayed section.

All parameters can be automated in your DAW or modulated within ZYKLØP [as explained here](#).



Every section has its own assortment of available modules. Some modules don't make sense for polyphonic material and hence are not available in the FX sections. Some are too CPU expensive to be rendered per voice and hence are not available in the OSC and FILTER sections.

# FX SECTION

The IRIS, OSC and FILTER sections are per voice. After the filter section the output of all voices gets summed and fed into FX section.



The SEND parameter determines how much of the signal is fed into the FX section.

## TIP

The SEND dials are PER VOICE. You can use this for example to modulate this parameter with an ADSR, such that the note onset is not sent into the FX to stay sharp and dry.

The OUT section has a LIMITER that can also be used for sound shaping.

The -6dB switch reduces the outgoing signal by -6dB.



## CAUTION

You should ALWAYS put a fast brick-wall limiter behind your synth when you are doing sound design to protect your ears!

# ARP SECTION

The ARP section processes the incoming midi data and offers a couple of midi-modules.

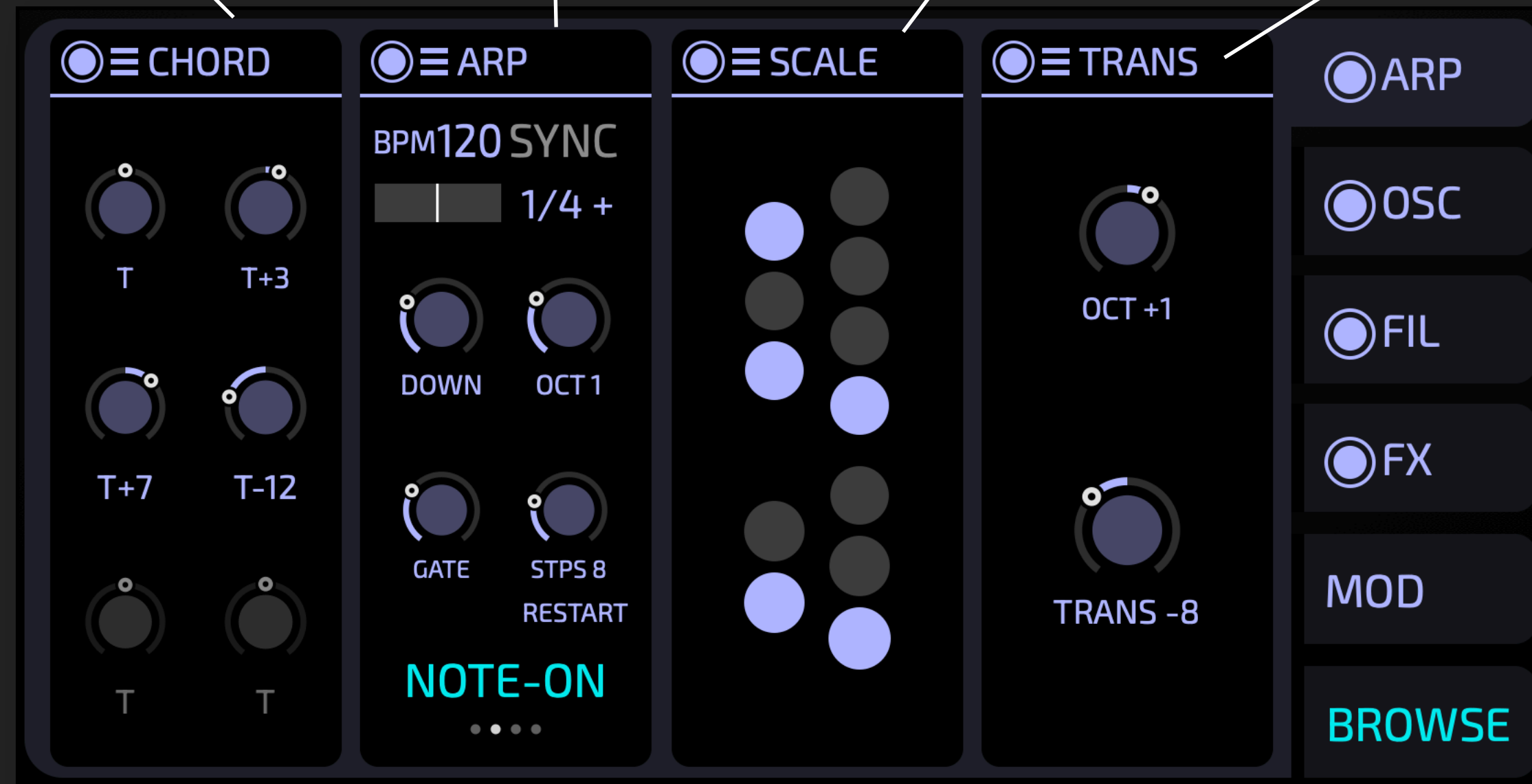
The CHORD module takes one input note and creates multiple transposed notes from it. For example, here the CHORD will produce the original incoming note T, one note 3 semi-tones above, one note 7 semi-tones above and one note that is one octave below the incoming note.

In this way you can play entire chords with one finger.

The ARP module takes all notes that are currently played (either on the keyboard or through the CHORD module) as an input for an arpeggio. GATE specifies the duration of the notes. OCT adds octavated copies. There are multiple modes to choose when the ARP should re-trigger.

The SCALE device allows you to choose which pitch classes are part of the scale. You can choose from plenty of pre-defined scales by clicking on the burger menu ≡. You can define your own scales by clicking on the circles.

The TRANS module can be used to transpose the midi-notes either by octaves or by semi-tones.



# MODULATE PARAMETERS

Modulating a parameter is really simple: click on the parameter to select it. The **corners** show that this parameter is selected.

Once a parameter is selected its name shows up on top of the modulators.

Now you can choose the depth of modulation for the selected parameter by dragging the circle next to the modulator.

In this example **LFO 1** has a positive influence on **CUTOFF**. The **KEYTRACK** has no influence, and high values of **MAIN ADSR** will reduce the **CUTOFF**.

You can **RIGHT-CLICK** on a modulator to remove it and to show all parameters that are modulated by this modulator.

SELECTED PARAMETER:  
**CUTOFF**

- E1 VELO
- E2 VELO
- E3 SLIDE
- MAIN ADSR**
- ADSR 1
- ADSR 2
- LFO 1**
- LFO 2
- LFO 3
- KEYTRACK

REMOVE

SELECT OSC1WAVE1MIX

SELECT FX1PHASER1FREQ

SELECT OSC1 POS

☰ LPF

RESO TRACK

**CUTOFF**

ACID

.....

Click and drag a **DIAL** or **SLIDER** to change the value. You can drag horizontally or vertically.

When a parameter is modulated the coloured ring indicates the current value. Every parameter can have as many modulations as you want. The modulations are added together.



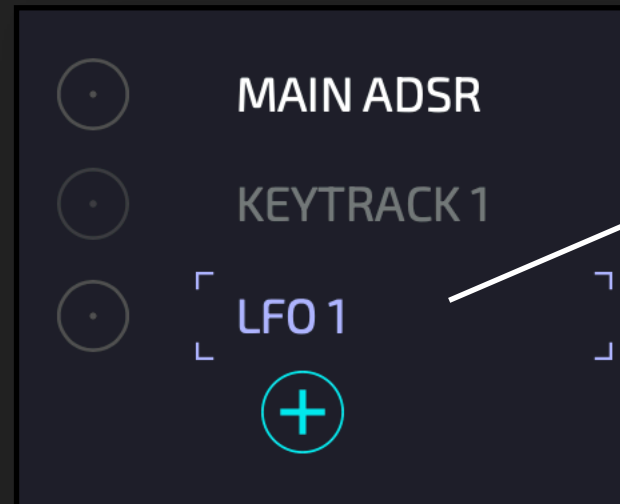
**TIP**

When holding **SHIFT** you can fine-tune values.

Double click resets the parameter to its default value.

The modulation sources are listed in the right sidebar of **ZYKLØP**.

# MODULATION SOURCES



Click to select a modulation source. You can now edit the settings in the detail page.

Click to choose from multiple pre-defined waveforms for the LFO.



You can also save and restore your LFO curves here as a "subpreset".



You can create and edit steps based on the grid with SHIFT-CLICK or RIGHT-CLICK.



With a few modulations you can turn any boring sound into something that sounds alive and interesting. Almost any parameter in ZYKLØP can be modulated.

You can also modulate the parameters of the Modulators. For example, you can modulate the rate of one LFO with another LFO. This allows you to setup complex, chaotic movements in your sound.

For special purposes you can add modules here. This works exactly like the "normal" modules - they modify the incoming modulation signal.

In POLY mode each note played has its own LFO. Otherwise there is one shared LFO for all notes.

Click here to choose a different grid for snapping.

# MODULATION MODULES

You can shape the modulation sources with these little modules. This can be very powerful - this is like CV modules in modular synthesizers.

**CURVE** allows you to shape the signals characteristic from linear to exponential or logarithmic.

With **SOFT** you can apply smoothing of the signal. You can set the time it takes to **RISE** the signal, and the time it takes to adopt to a **FALL** of the incoming signal.



TIP

You can chain multiple **JIT** modules with different settings to create unique random patterns.

**JIT** adds random fluctuations to the signal. You can set the **RATE** of how often the random value changes and the **AMT** of randomness added to the incoming signal.

**NOTE-RANDOM** creates a new random value at note-on and holds it for the entire note.

**VISU** is a visualizer for the modulation signal. Each note's signal is represented with a dot.

**AMT** simply scales the amplitude of the signal.

● ≡ AMT   ● ≡ CURVE   ● ≡ SOFT   ● ≡ JIT   ● ≡ RND   ● ≡ VISU

AMT   CURVE   RISE   [RATE]   ●   AMT

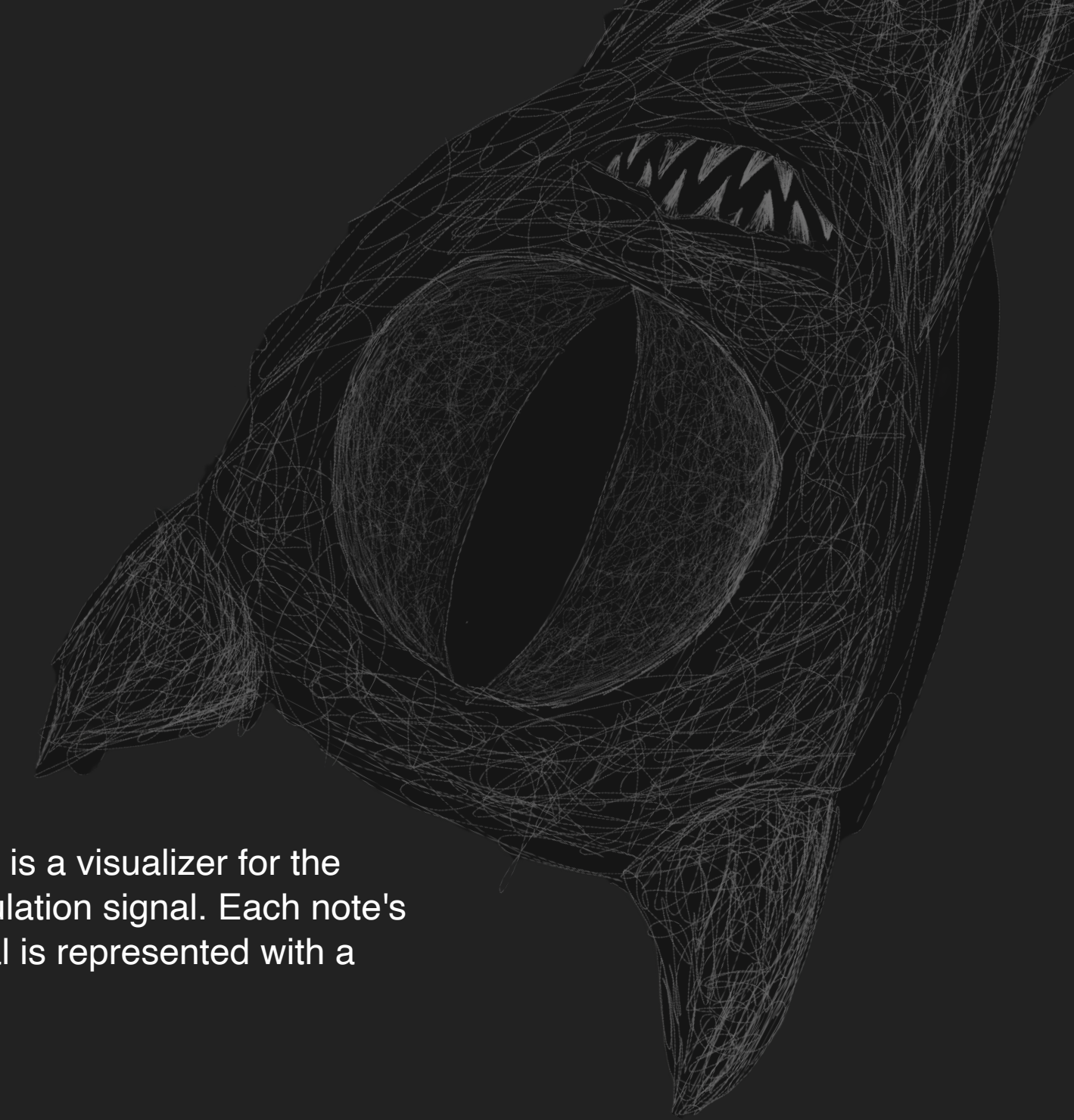
FALL   AMT

● ≡ POS

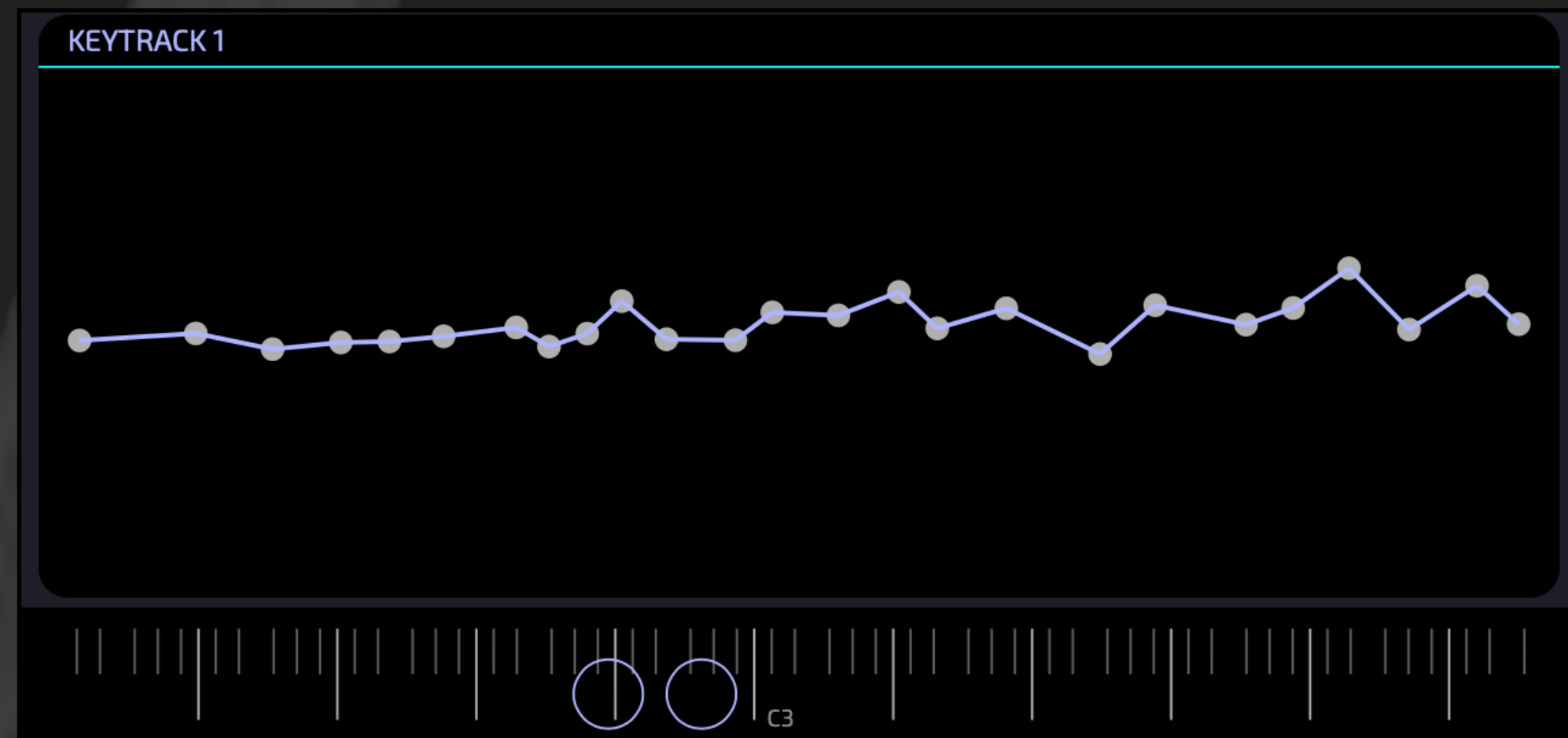
ROTATE

OFFSET

**POS** is an "extension" module for the **LFO**. It allows you to add an offset to the **LFO** position. Of course, this **OFFSET** can be modulated, e.g. by the **LFO** itself ... there are no limits.



# MODULATION: KEYTRACK



**KEYTRACK** takes the note input and uses a curve to determine the output signal.



In the example I draw a random curve around center. This can now be used to modulate **DETUNE** of the **OSC** - in this way every note has its own slight imperfections. Techniques like this can breathe a lot of life into your sounds.

# MODULATION MATRIX

Sometimes it's useful to see an overview of all modulations in the preset. For this purpose click on MOD to bring the MODULATION MATRIX up.

You can adjust the modulation depth directly here in the matrix.

This column shows the parameter that gets modulated. You can click the parameter to "jump" to the parameter in the UI and select it.

This column shows the section that the modulated parameter belongs to.

This column lists the modulation sources.

Click on the X to remove a modulation.

If the modulated parameter is from a module, then the module name is showed here.

REMOVE	MOD SOURCE	DEPTH	PARAM	SECTION	MODULE
×	LFO 1		POS	IRIS	
×	E 2		STRING	IRIS	
×	E 3		SYNC	IRIS	

ARP  
OSC  
FIL  
FX  
MOD  
BROWSE

Click on MOD to bring up the modulation matrix.

# MODULE: ANALOG FILTERS

There are multiple filters that use virtual analog modeling.

The **LPF (LOWPASS)** module has multiple classical low pass modules with very different sound characteristics ranging from clean to dirty and screaming.

**TRACK** turned fully right shifts the **CUTOFF** frequency with the note you play.



**TIP**

Put the filter into self-oscillating mode by using high resonance to use it as a generator. Turn **TRACK** fully right. To tune, add a sine generator after the filter (**WAVES** module) and tune the **CUTOFF** by ear.



This is a classical 4-pole ladder design that can be configured to work as a lowpass, highpass, bandpass or bandstop (notch) filter.

It can be run in a dual configuration with two filters either parallel or serial. The **OFFSET** parameter determines the difference between the two **CUTOFF** values.

# MODULE: COMB and RESONATOR

The RESONATOR module uses a so called "modal filterbank" - 32 bandpass filters in parallel. This filterbank can be "trained" by any sample via drag-and-drop.

A comb filter is a specific kind of resonator, often used for classic physical modeling.

**HARM** influences the inharmonics in the signal.

**DAMP** lowpass filters the feedback path.

**FB** sets the amount of feedback. It can be positive or negative feedback, which has a different sound.

There are two modes: **CLASSIC** is the normal mode. **PHYSIS** has a different tonal character, and the **FB** behaves differently.

**PITCH** transposes the entire filterbank lower or higher.

**RESO** controls how easily the resonators respond and how long they "ring".

**RESO** is a resonator with a *fixed pitch*. This is great to add character to a sound



**TIP**

The fun starts when you use your own samples to train the modal filterbank - you can simply drag-and-drop a sample. ZYKLØP will derive a "fingerprint" that is saved with the preset, access to the source sample is not needed.

**TRACK** can be set to full - and the filterbank will follow the pitch of the incoming notes

**INHARM** can be used to add inharmonics to the sound.

Click here to train the filterbank to a specific sample.



DEFAULT
CHOOSE WAV ...
AFRICA
AMBIENT
CRYSTAL_GLASS
GLOCK
HANDPAN
HOF

# MODULE: WAVE

The WAVE module generates classic "vintage" synth waveforms with up to 5x unison. It acts like an analog oscillator

You can **DETUNE** the unison voices. For example you can use this to create super-saws.

**UNI** chooses the amount of unison voices.

**MIX** blends in the generated wave with the original signal. For example you can chain 2 WAVE modules for 10x unison.

You can choose the waveform: SAW, SQUARE, TRI, SINE and NOISE.



You can **SPREAD** the unison voices in the stereo panorama.

**OCT** transposes some of the unison voices by octave(s)

**RAND** randomizes the phase at note-on. Activate this to avoid a "zip" like start of the wave.

**PM** is pulse-width modulation for the **SQUARE** wave. For **TRI** and **SINE** it distorts the phase to make the waveform more sawlike. For **NOISE** it is a temporal bit-crushing effect.

You can transpose the oscillator by  $\pm 2$  octaves.



Use two WAVE modules and one LPF module, and you have a two-OSC analog modeled synth.

You can use the IRIS to add dirt and grit to make the sound more organic.

You can also create your own super-saw by using one or two WAVE modules and high unison.

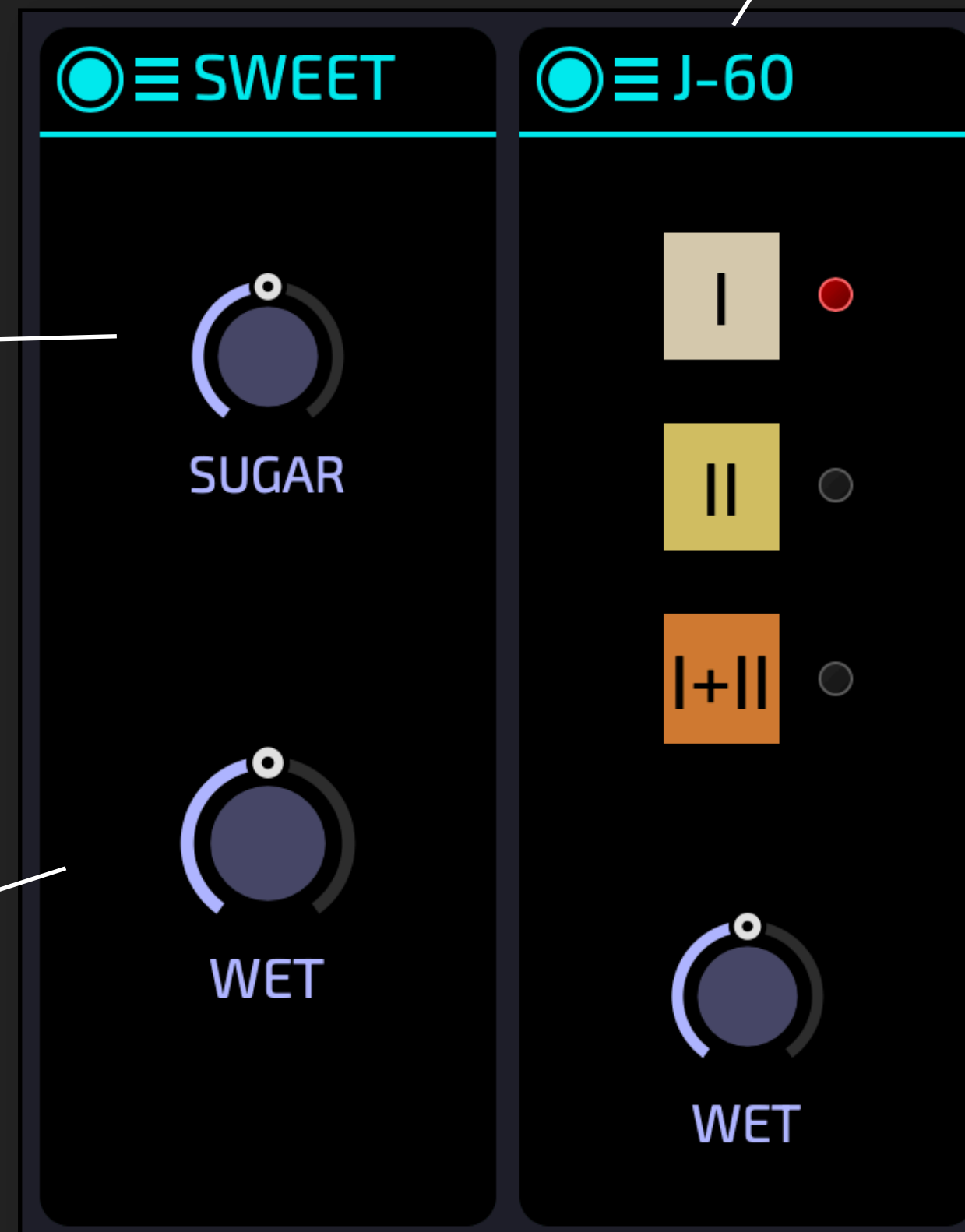
# MODULE: SWEET / J-60

These modules can be used to add a subtle or not so subtle chorus effect.

**SWEET** will make most sounds just ... sweeter. Used with subtle amount it can give a very modern and velvety sound.

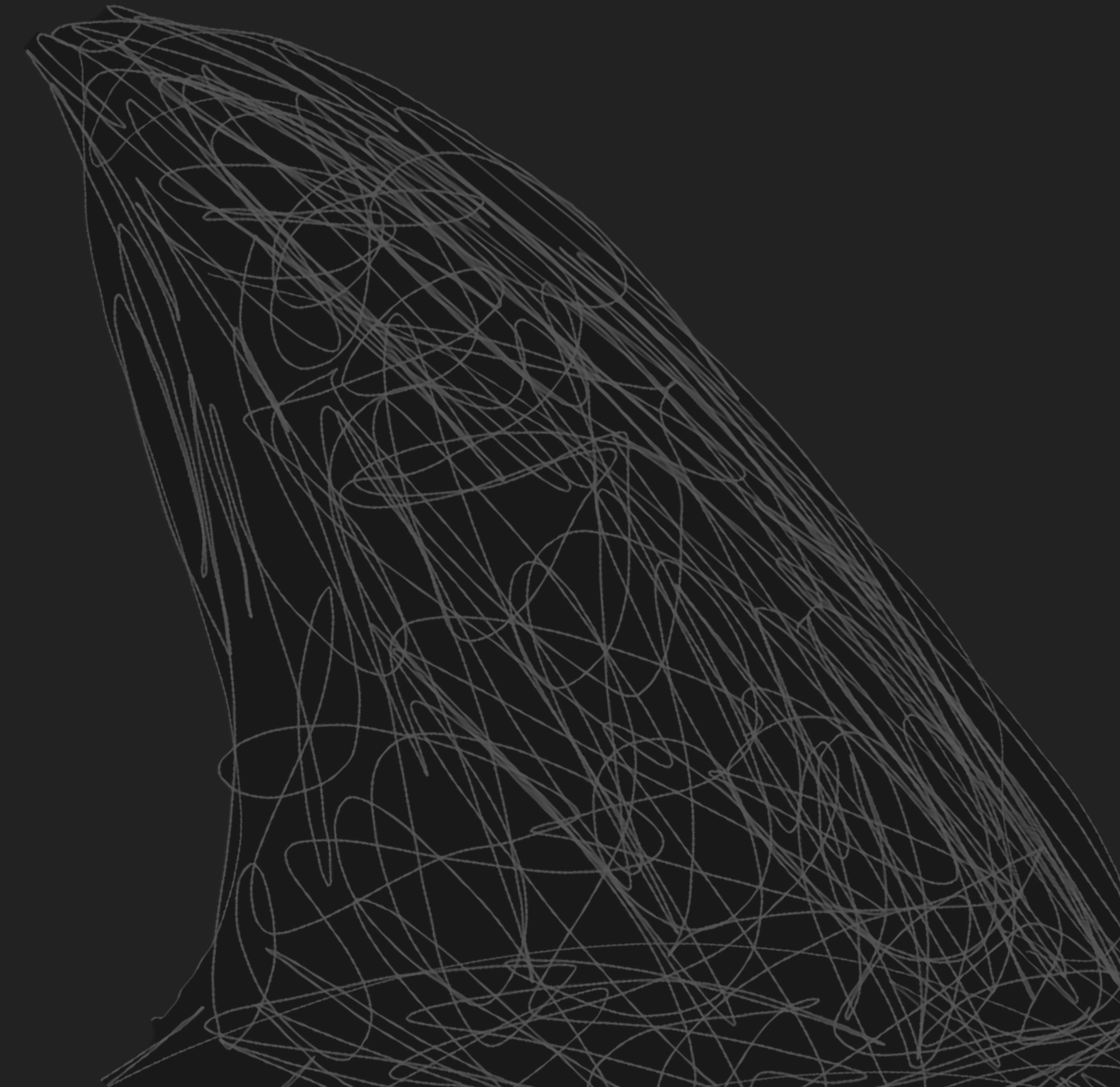
High values of **SUGAR** create Ligeti-like detunings for an eerie or haunting sound character.

Simply the amount of the effect applied.



The JUNO 60 is an iconic vintage synth from Roland ([Wikipedia](#)). It contained a special chorus effect which became its signature sound. The **J-60** module in **ZYKLØP** is a virtual analog model inspired by this chorus.

It is a cheat-code, as it can make almost everything sound better.



# MODULE: CLOUDS + REVERB + SHIMMER

There are three different types of REVERB modules available.

**CLOUDS** is inspired by a reverb plugin called *CLOUDSEED* by [Valdemar Erlingsson](#). It's great for epic ambient reverbs with long or even endless tails.



TIP

If you find a nice setting you can save / restore it via the burger menu ☰

**REVERB** is a simple vintage reverb inspired by the 80s algorithmic reverbs.



**SHIMMER** is a shimmer effect. It uses a pitch shifter within a feedback loop and can create angel-like clouds of sound, but also dark or dissonant textures.

# MODULE: LOOPHOLE

LOOPHOLE consists of six parallel micro-loopers with incommensurable delay times.

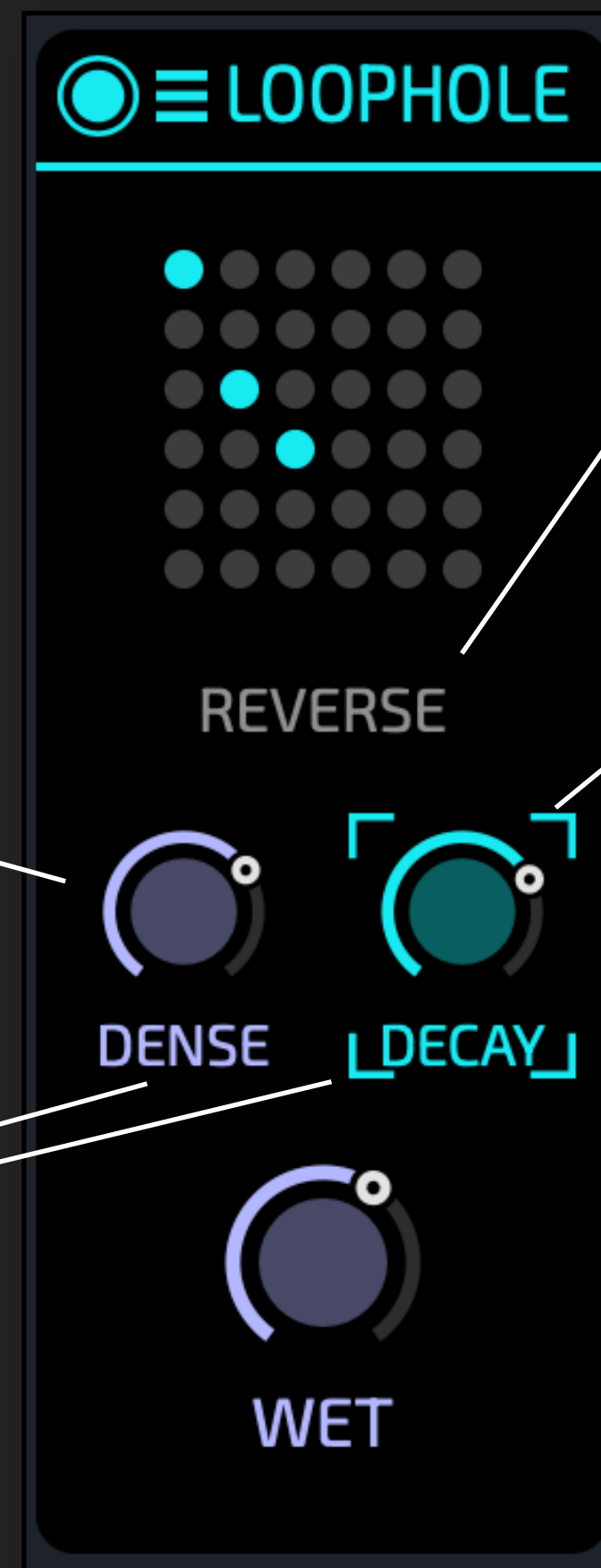
It's very much fun ... you can take something plain (like a sine), and use a LOOPHOLE module. Just by playing a few notes and tweaking some transformers with the mouse you will get interesting sound textures.



LOOPHOLE is inspired by a plugin called [Weeping Wall](#) by [Aqeel Adam](#), which has much more functionality and works great on synths and keys as well as on other acoustic instruments like guitar.

With low DENSE you get stuttering echoes.

With high DENSE and DECAY you create very organic "freezes"



Activate for REVERSE playback.

DECAY on full will loop forever, but new audio material is also slowly fed into it



# MODULE: GRAINS

This is basically the *SWARM* mode of the plugin *LOVE* as a module.

**GLITT** sets the probability that a grain is pitched an octave up (turn right) or down (turn left). In center position all grains are played back at their original octave.

**JIT** adds randomness to the grain creation. With low values the grains play at a steady rate, with high values it is chaotic.

Simply the amount of the effect applied.



Each grain gets its own tuning. This parameter controls how much the tuning is offset. High values create Ligeti-like clouds of dissonances.

**CALM** controls the grain duration from very short to very long and also the grain density. Turn fully right for smooth sound and long tail.

# MODULE: ORBIT

ORBIT is a new type of modulation effect. It shares similarity with (analog) ring modulation and frequency modulation, but has its own unique character.

Sonically behaves in a similar way as the **RATIO** of classical FM.

Technically it controls the ratio between the frequency of the moon orbits to the frequency of the planet orbiting the sun (see explanation to the right)



**FLUX** can be used to add inharmonics to the sound.

Technically this works by adding a fixed frequency to the moon's orbiting speed - this desynchronizes the orbits of the celestial bodies.

**CHAR** - the ORBIT effect is most pronounced in center position. There are two different ways to tone down the effect by either turning right or left.

Technically this influences the ratio between the sun-planet distance and the planet-moon distance.

## HOW IT WORKS

If you happen to own the KULT synthesizer you can use a sine OSC and then apply FM or AM. You will see that this transforms the signal trajectory by adding curves and spirals.

The orbit modulation interprets the incoming signal as a planet orbiting the sun. It then adds a moon to this planet and follows the moons trajectory. This new trajectory is then transformed back to sound.

Like a real moon, its orbits around the planet can be circular or elliptical, as well as centered or off-center. This can be chosen with the modes.

There are three different modes - each with its own sonic character. You don't need to understand the theory behind this, but it's explained here.

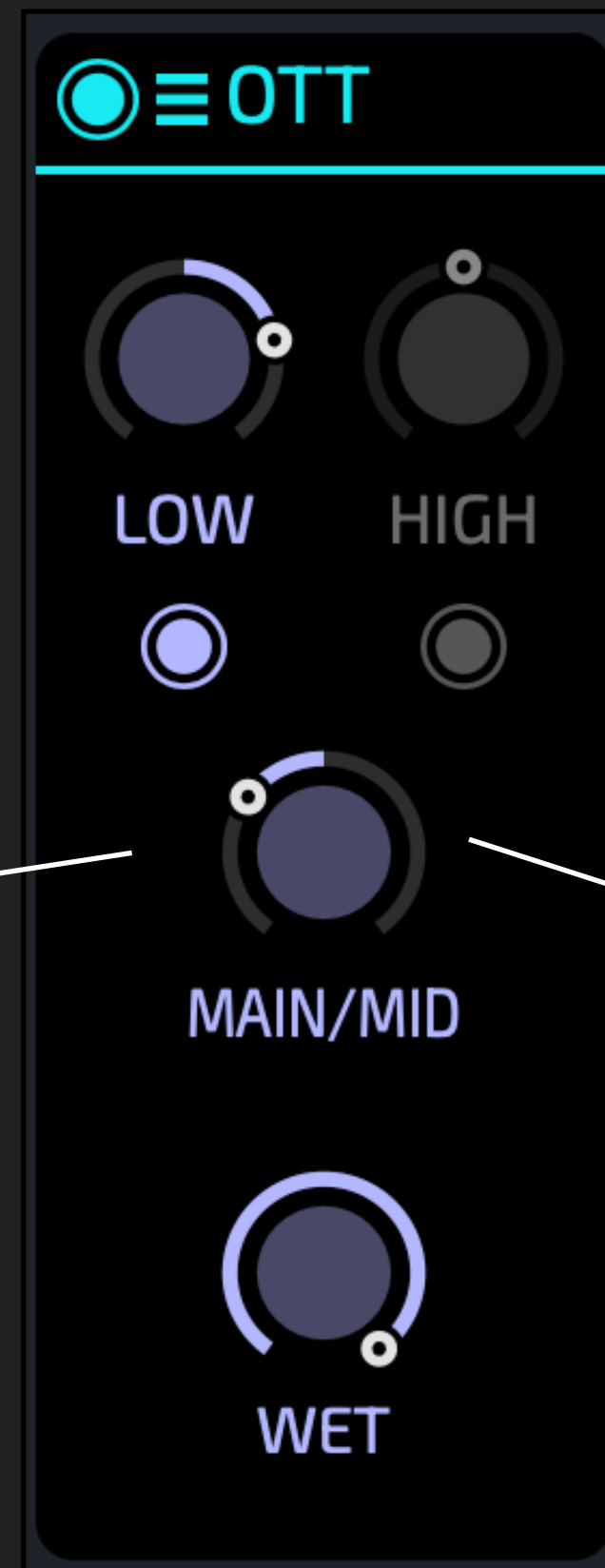
# MODULE: OTT

OTT stands for Over-The-Top and has become a standard technique in modern electronic music.

This module is a multiband-compressor and can be used to add texture, punch or saturation to a signal.

Its available in the OSC and FILTER section - hence you can use this per voice.

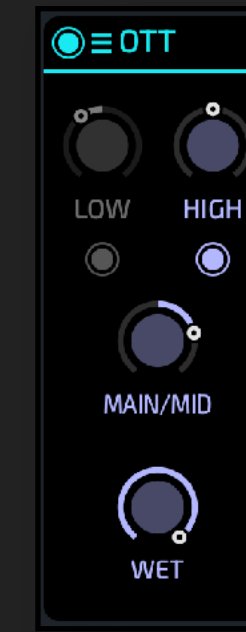
Turned to the **LEFT** this is (aggressive) upwards compression: silent signals are made louder. This can be used to bring up texture in the sound or bring out silent parts like tails.



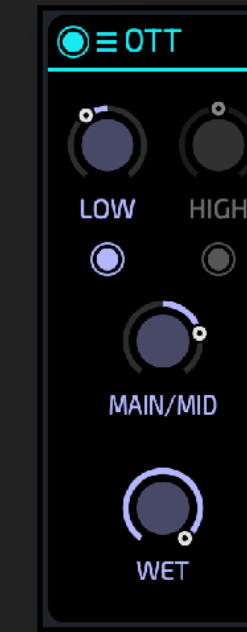
Turned to the **RIGHT** this is "normal" downwards compression: loud signals are reduced to compress the signal. Gives punch.



MAIN/MID applies to the entire frequency range.



MAIN/MID applies to the low end and mid range. Separate dial for high range.



MAIN/MID applies to the high end and mid range. Separate dial for low range.



Separate dials for each of the bands.

## ⚡ CAUTION ⚡

When you turn a dial to the left it is doing upwards compression. That is: silent signals get dramatic amplification. Be careful with very silent tails when you put the OTT in the FX section.

# MODULE: FM

FM uses the incoming signal to modulate a low frequency sine wave. This can be used to create metallic and inharmonic tones, to add sharp overtones, or to animate a static signal.

The incoming signal modulates a low frequency sine wave. The frequency of this sine wave can be chosen here with a wide range from ultra-slow to far into the audio range.

Low values animate the signal. Higher values create inharmonics.



SPREAD can be used to treat the LEFT and RIGHT channels differently. This can range from subtle to wild.

The amount of frequency modulation that is applied to the incoming signal.



To best understand this module feed it with something simple, e.g. a sine wave. You can use the INIT preset for this.



Unless you are seeking a rough and sharp sound chaining a J-60 chorus or a SWEET module can work wonders!

# MODULES: RM

The modules RM and can be used to create inharmonic, metallic sounds, or to add motion to a sound.



To best understand this module feed it with something simple, e.g. a sine wave. You can use the INIT preset for this.

RM modulates the incoming signal with a sine wave. The frequency is set with the **RATIO**.

RIGHT-CLICK to choose from some integer ratios.

1/3	-1/3
1/2	-1/2
2/3	-2/3
1	-1
2	-2
2.5	-2.5
3	-3
4	-4

There are two modes available: **RING** and **BELL** that have a very different sound character.

RM stands for Ring-Modulation



FLUX adds a fixed low frequency to the sine. It can be used to create a metallic sound.

# MODULE: AM

This offers amplitude modulation. It can be used to add inharmonic character or animation to a sound.

The AM is tuned to the incoming signal. With **RATIO** you can choose the frequency ratio between the incoming signal and the signal used for modulation.

The amount of the amplitude modulation.



Normally the **RATIO** of an amplitude modulation creates harmonic tones for integer ratios, and inharmonic values in between. Hence sweeping the **RATIO** gives a pretty wild journey through all kind of inharmonics and is often of limited musical use.

This module uses the underlying technique of [shepard tones \(Wikipedia\)](#) to allow for a separation: with **RATIO** you determine how high or low the modulation sounds, with **HARM** you control the specific timbral character of the in-harmonics.



To make yourself familiar with this module start with the INIT sound, or keep all the **TRANSFORMERS** to their defaults and dial up **BLUE** to the max.

Once you have an idea how **RATIO** and **HARM** influence the sound start experimenting with the colour-transformers.

# MODULES for Saturation / Distortion

These modules can be used to distort and shape the sound.



These modules can be used to shape the sound, add punch, or more overtones. All parameters can be modulated. Often these create interesting effects if used in conjunction with other modules.

ASYM offers asymmetric distortion.

CRUSH offers two sorts of bit-crushing.

SATURATOR has multiple different distortion types. You can set the DRIVE.



Use ATT to attenuate the saturated signal, before the DRY / WET.

# MODULE: TRANS

This module is only available for OSC section It can be used for tuning the IRIS *and* the modules in the OSC section.



TIP

This module also affects the IRIS, hence it will always be in the left-most position of the section.



TIP

In many cases you just want to put the IRIS up or down in octaves. You don't need the TRANS module for this - just use this dial.

Tune the oscillator up or down in cents of a semitone. Modulate this with a **PERLIN** noise modulation source for sweet detuning.

Tune the oscillator up or down by octaves. It's fun to modulate this, eg with an LFO.



Transpose the oscillator up or down in semi-tones.

If active the **TRANSPPOSE** snaps to integer semi-tones.

If active the oscillator always plays note C3, regardless of the note you play. You can use the **TRANS + OCT** to change this note.

# MONO MODE

The MONO MODE can be used for basses and leads that have similar behaviour to monophonic analog synths.



You can drag and drop the last played notes as audio from here.



TIP

You can play a note and then drag this as audio from here to the IRIS - to get the sound re-synthesized!

Normally ZYKLØP is a polyphonic synth. By activating MONO you can make it a monophonic synth with legato function.

With TIME you can control the legato speed.

When GLISS is active you will also have a glissando between two notes if you play legato.

Incoming notes are displayed on the virtual MPE keyboard display.

The size of the outer circle line indicates (polyphonic) PRESSURE or AFTERTOUCH.

The filled circle area indicates the VELOCITY.

# CPU Performance

ZYKLØP has been designed and optimized with great care for CPU efficiency.

On the other hand, a single preset can have:

- IRIS with 10 transformers and 4 layers
- 6 modules which are per voice
- 4 midi effects
- 3 audio effects
- 10 modulation sources
- ...

It should not come as a surprise that this gives you some possibilities to strain your system.



≡ SETUP OSMOSE\*

MASTER	440
GLIDE	12
PB	2
CPU	ECO

A screenshot of the ZYKLØP software's setup menu. The menu is titled '≡ SETUP OSMOSE\*'. It lists several settings: MASTER (440), GLIDE (12), PB (2), and CPU (ECO). A white line points from the text to the right towards the 'CPU' setting.

You can set the CPU mode here. This affects a lot of internal settings, but in most of the cases the effect will be subtle. **ECO** and **SMART** mode will be sufficient for almost all cases.

⚡ IMPORTANT ⚡

When the CPU usage is problematic always check the voice indicator first.

Long, exponentially shaped release tails are a recipe for high CPU strain.

# PRESET PACKS

You can create PACKS from your presets and other users of ZYKLØP can install these.

PACKS are a great way to extend the sonic capabilities of your instrument.

You can simply drag any .ZyklopPack to the ZYKLØP user interface to install the pack, or choose **INSTALL PACK** from the main menu.

Installed packs shows up here



It is easy to create your own packs. Simply use the file system and create a folder that contains ZYKLØP presets. You can organize these in a single level of sub-folders, deeper nesting is ignored.

Choose **CREATE PACK ...** from the main menu. You will only need to locate the folder with your presets and a name to save the pack.



.ZyklopPack

DROP TO INSTALL PACK



If you handle preset files, you should ALWAYS make a copy beforehand.

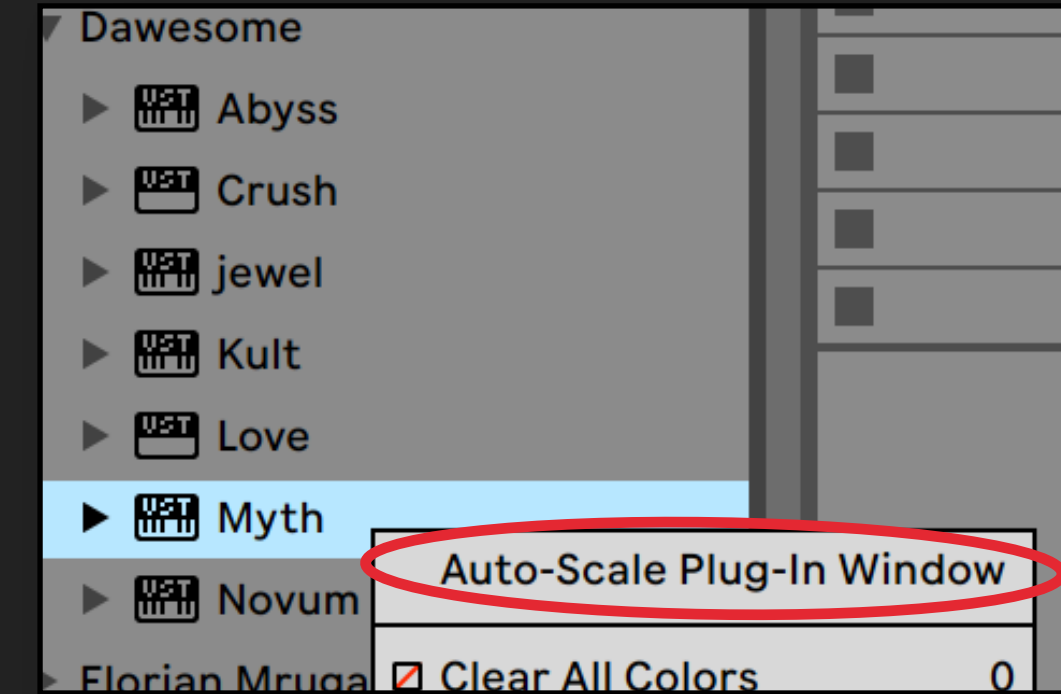
Better safe than sorry!

# FAQ / Troubleshooting

Q: The UI looks pixelated / distorted

A: In the main burger menu in the upper left there is an entry to [USE OPEN GL](#). Please try to de/activate this setting, then reload your project - the plugin needs to be reloaded to make this change active.

If you are using [Ableton LIVE](#): make sure that “Auto-Scale Plug-In Window” is **NOT ACTIVATED**.



Q: I can't hear anything!

A: First make sure that [ZYKLØP](#) receives midi data - you can see incoming midi on the virtual keyboard. Make sure that the OUT is not set to zero.

Q: I installed [ZYKLØP](#), but it does not show up in my DAW?

A: In most DAWs plugins are listed by manufacturer name. You find [ZYKLØP](#) listed under [DAWESOME](#).

If this does not help - please make sure that VST3 (or AU) is activated in your DAW.

For [Pro Tools](#) users: plugins come in various formats; the most common are VST, VST3 (and AU on Mac) - these are the industry standard. Pro Tools has a proprietary plugin format and is not able to host VST or AU plugins. [ZYKLØP](#) is available only as VST3 and AU. But there is a workaround: you can use a wrapper plugin. There are two that I can recommend:

- [KushView Element](#) - this one is even free
- [BlueCatAudio Patchwork](#)

You load the wrapper Plugin in Pro Tools, and then you insert [ZYKLØP](#) in the wrapper - it is simple and works as if you had opened [ZYKLØP](#) in Pro Tools itself. The advantage is: there are many plugins on the market that are available only as VST or AU - with this wrapper you can use them all in Pro Tools.

# FAQ / Troubleshooting

Q: I am experiencing audio drop outs - what a lousy plugin!

Q: I have a question / feedback - where can I leave it?

Q: I have a cool idea for a great feature!

A: Just drop me an email to [peter@dawesomemusic.com](mailto:peter@dawesomemusic.com) - I appreciate any kind of constructive feedback and I am trying my best to have any user satisfied, regardless whether you purchased or not. Usually I try to answer within a few days.

If you want to share any idea with me please drop me an email to [peter@dawesomemusic.com](mailto:peter@dawesomemusic.com). Please note that I may have had the idea before and hence I won't pay you license fees if I choose to implement this idea or a related idea in one of my plugins. If you believe your idea has tremendous commercial potential make sure to get a signed agreement *before* sharing the idea with me / anyone.

Q: I like your work - how can I support you?

A: Thank you - this is my real reward for the work I am doing! I hope you will find lots of fun and inspiration with [ZYKLØP](#) or any other of my plugins. If you want to support me: spread the word - many (most?) people simply have not heard about DAWESOME.

Or send me a brief message via mail to [peter@dawesomemusic.com](mailto:peter@dawesomemusic.com) - this will certainly cheer me up!

# CREDITS - THANK YOU!

- [ZYKLØP](#) is implemented in C++ using the [Juce Framework](#). I am grateful for its existence and for the community of JUCE developers.
- [Valdemar Erlingsson](#) is the creator of the gorgeous free reverb plugin called [Cloud Seed](#). I took inspiration from his work for the CLOUDS FX
- [Nigel Redmon](#) has published an intriguing [series](#) about analog ADSRs. I took inspiration and design choices from his series.
- [ZYKLØP](#) uses the awesome AVIR image resizing algorithm designed by [Aleksey Vaneev](#) of Voxengo
- Sample rate converter designed by [Aleksey Vaneev](#) of Voxengo
- [Rich Whitfield](#) and [Frank Gesang aka SiL3NC3](#) and [Saf Ro](#) and [DATABROTH](#) painstakingly proof read this user guide
- [DATABROTH](#) + [Spektralisk](#) + [HydraTek](#) + [Rich Whitfield](#) have provided extensive feedback / discussion of various features over a long time
- [Sound Author](#) + [Sabastian Weaver aka Azure Eyes](#) gave feedback to improve features and had some very cool feature requests
- [Sound Author](#) has created and provided most of the IRIS samples
- [BIANSU](#) created the official trailer track for MYTH, which is the big sister of ZYKLØP
- [Jacky Ligon](#) and [Andreya](#) are tremendously helpful. They convinced me to add MTS-ESP support for the bigger sister MYTH
- [Jacky Ligon](#) and [Sound Author](#) have provided samples for the modal filter / resonator
- [KiroArt5](#) has made the charming Zykløp drawings

I am blessed with an awesome group of [Beta Heroes](#) who found bugs and provided many ideas:

[Rich Whitfield](#)  
[Cool WAV](#)  
[Tomavatars](#)  
[sadà\exposadà](#)  
[AstralMuse](#)

[DATABROTH](#)  
[Ei°HYM](#)  
[Andreya](#)  
lab by the sea  
dreamerOnGo

[Spektralisk](#)  
[Saf Ro](#)  
Squaremoons  
David Henkel

[HydraTek](#)  
[Florian Mrugalla](#)  
Mathias Brüssel  
[BIANSU](#)

[Sound Author](#)  
[HiEnergy](#)  
Philip Rampi  
[Jacky Ligon](#)

[Alessandro Cardinale](#)  
[Frank Gesang aka SiL3NC3](#)  
[David Lilja \(PaleSkinnySwede\)](#)  
[Sabastian Weaver aka Azure Eyes](#)

# ABOUT DAWESOME



Hey,

I'm Peter, the creator (and chief mad scientist) behind Dawesome plugins.

By day, I'm deep in the code, crafting synths and effects to inspire creativity. By night, you'll find me playing bassoon and contrabassoon in a few ensembles — and bringing my lifelong love of synthesizers into everything I do.

With a PhD in math, I've earned my "professional nerd" badge, and I wear it proudly

Here are a few principles that keep me going:

- \* Creativity should be fun — If it doesn't spark joy, let's tweak it till it does.
- \* Sense and Simplicity — The simplest ideas often come from the toughest work
- \* Exploration is the royal road — There's no map; we're here to discover.

Thanks for being part of the Dawesome journey!

Peter ✓

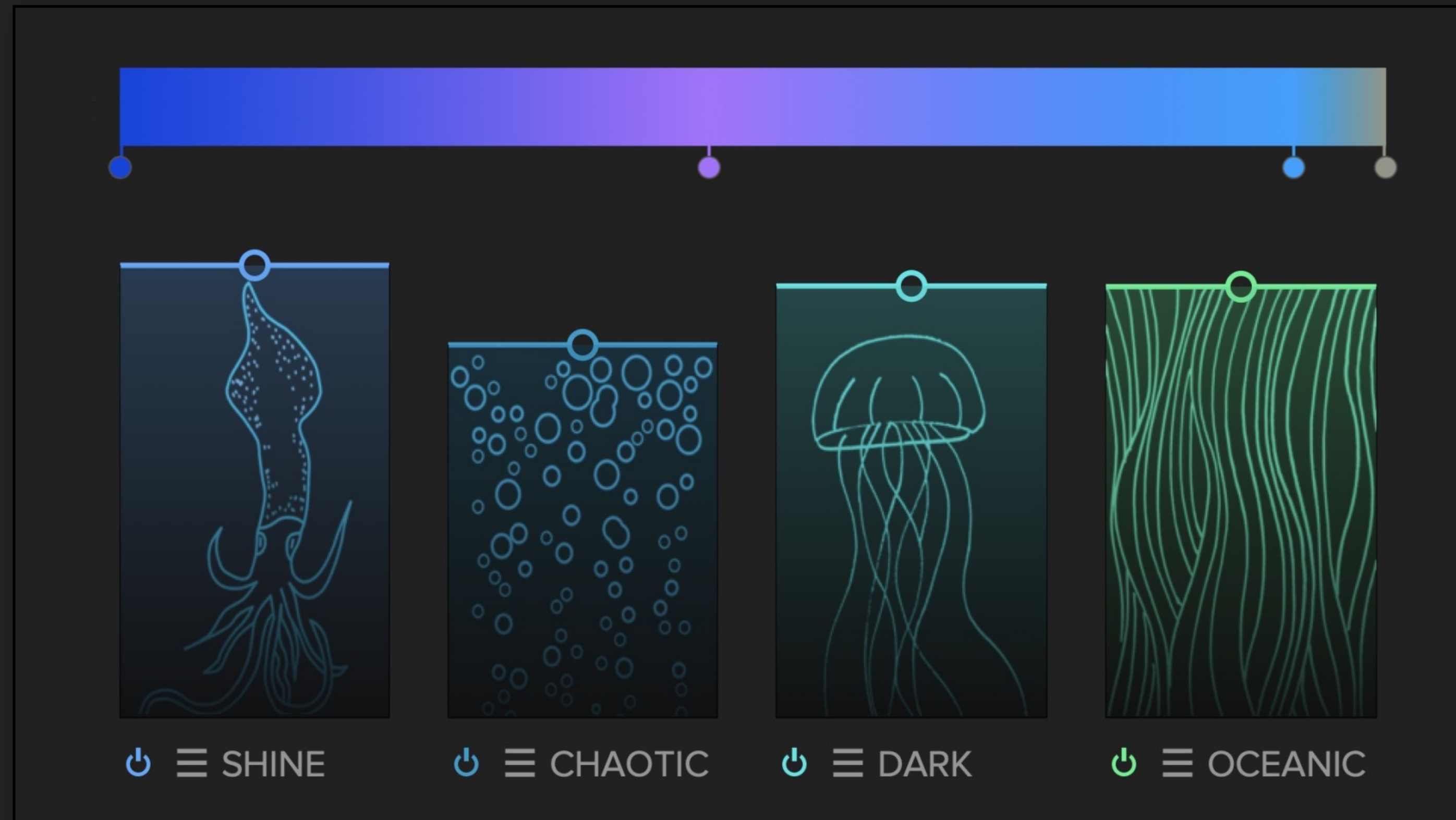


DAWESOME

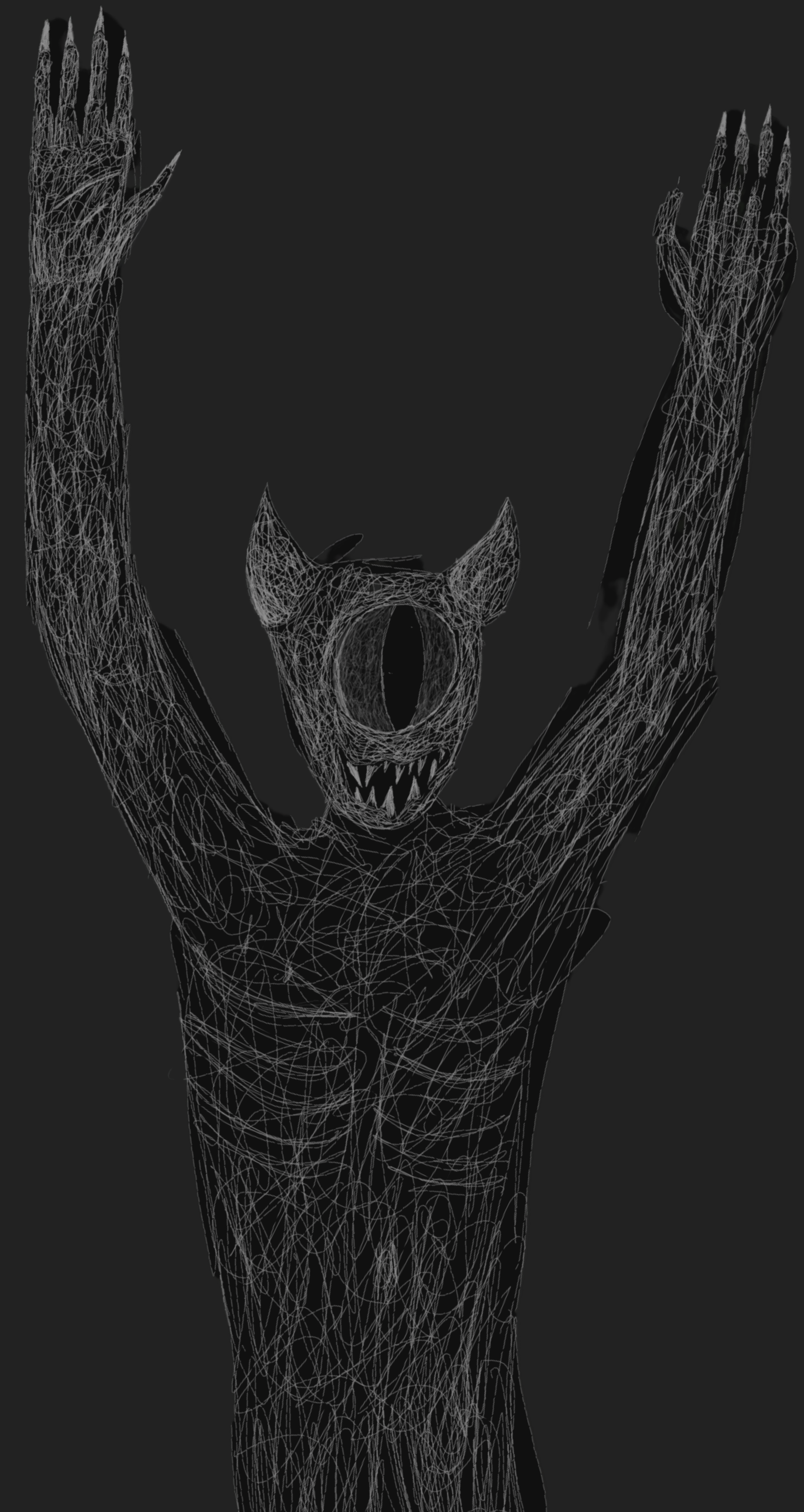
# Dawesome Village

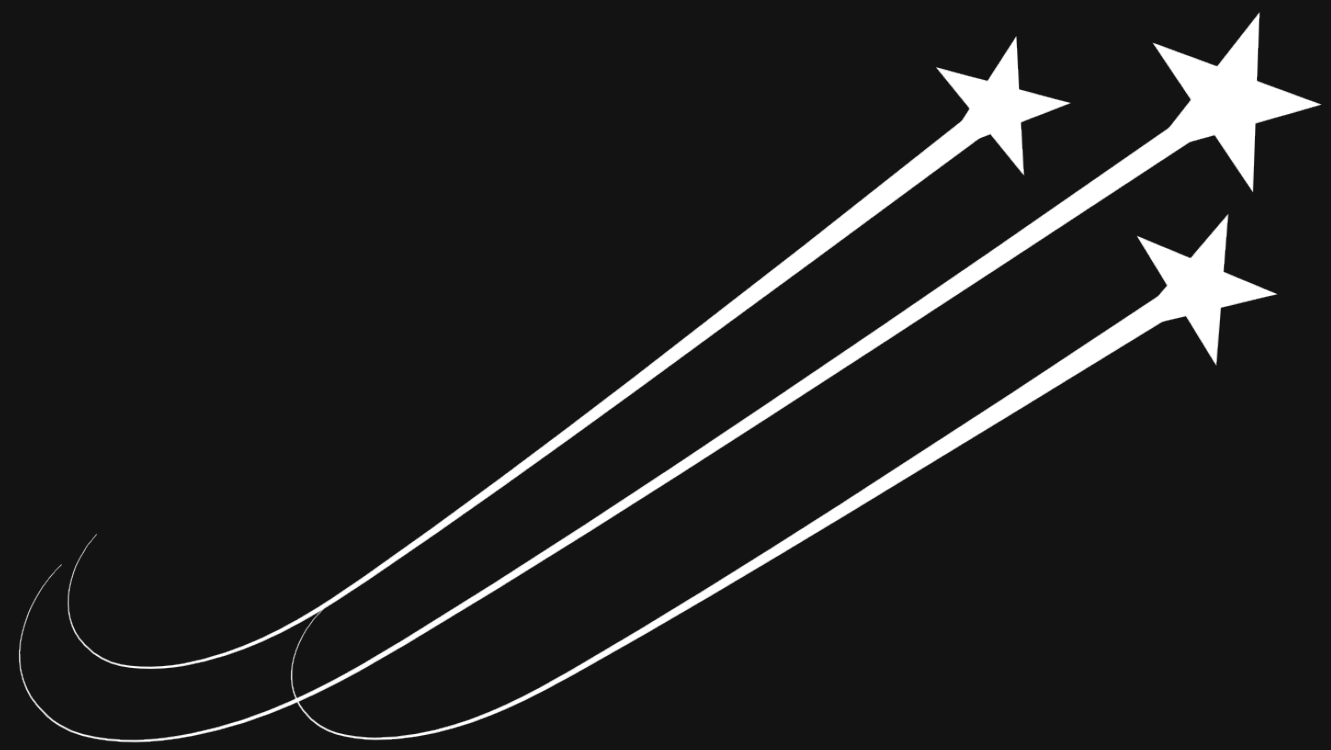
... is the **friendly and supportive community for sound lovers** on Discord!

You can join by clicking on the invitation link below. You will find many interesting people, tons of information, many freebies like sample packs and presets and much more...



Invitation Link: [Dawesome Village](#)





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