

DiBiQuadro



Mercury2CP

USER MANUAL



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Thanks for purchasing DiBiQuadro Mercury2CP. This is the result of a meticulous engineering design and passion for perfection.

Please take your time to read carefully the content of this manual before using the plugin and the renderer.

Master of Distortion

Mercury2CPU is a cutting-edge simulation that captures the spirit of a classic amplifier revered by guitarists worldwide, offering a rich palette of tones suitable for a myriad of musical genres.

The amp's Rhythm channel boasts a pristine, glassy quality that sparkles with warmth and depth, ideal for genres such as jazz, country, and ambient music. The Lead channel unleashes a torrent of harmonically-rich saturation, characterized by tight lows, soaring mids, and shimmering highs. From classic rock to progressive metal, the Lead channel of Mercury2CP empowers guitarists to sculpt their sound with precision and authority, offering a seamless transition from creamy bluesy tones to aggressive, high-gain saturation. Its versatile EQ controls allow players to sculpt their tone to perfection, while the responsive dynamics ensure a musical playing

experience that is both inspiring and exhilarating.

In terms of sound quality, Mercury2CP stands head and shoulders above the competition, with meticulously crafted algorithms that faithfully replicate the nuances of the original amplifier. Mercury2CP captures the essence of its analog counterpart with unparalleled authenticity and fidelity.

Non-Plus-Ultra Preamp: the perfect approach to circuit modelling

Mercury2CP preamp features a built-from-scratch 64-bit engine which is based on general purpose hi-quality algorithms and the most recent non-linear models. Particular attention has been put on the definition of the Non-Linear Solver to get the maximum quality and realism, without losing details in favor of performances (Double-Precision Non-Linear Computing).

Also, Mercury2CP brings a completely new philosophy to the table: being able to shape your tone by giving total access to the circuit components. Managing each gain stage level, shaping the tone-stacks or modifying the tubes characteristics: there is virtually no limit to what you can achieve with Mercury2CP!

Pwr/Cabs: Non-Linear Convolution

The chain made of tube power amp, cabinet/speaker, microphone, high-end studio preamp is sampled together as a unique block to preserve the correct interaction between the specific power amp and the speakers. Contrary to what is commonly used in the amp simulation market we use Non-Linear Convolution (NLC) to capture the harmonic distortion of the power amp, the speaker break up, and the harmonic distortion introduced by the high-end mic preamp.

This is not possible using the commonly spread approach of a single IR. First, with a single IR is impossible to reconstruct the non-linearities of an analog chain. Since the power amp is generally modeled in the competitors products, the main problem is the missing non-linearities coming from the speaker and the microphone preamp. While this is somehow negligible for some types of high wattage speakers when dealing high gain sounds, it is essential when the speaker break up is part of the tone. Second, most of the times a solid state power amp is used to sample a speaker behavior. This power amp despite being considered "neutral" has its own character due to the different impedance and interaction with the speaker compared

to the tube amp we generally want to capture.

The NLC engine, based on 64-bit (double precision) processing allows up to 40 kernels (harmonics). The amount of kernels is selectable.

Another limit of many well-known products on the market is the complete absence of power amp clipping. This aspect is not only provided by our Pwr/Cabs but a complete sets of parameters allows the full control over the power amp headroom, the type of clipping and the interaction of the negative feedback with the Depth and Presence knobs. This is essential to determine the amount of signal used in the tone shaping after the power amp starts to saturate. The power amp saturation adds harmonics to the kernels already applied by the NLC engine.

DNC+

The proprietary solutions described in the previous paragraphs are grouped under a common acronym: DNC+ (Double-Precision Non-Linear Computing and Convolution).



Double-Precision Non-Linear Computing and Convolution

About (db)²

DiBiQuadro is the result of the joint forces of two Italian Software Engineers with more than 15 years of experience in several industries including Biomedical, Military, Insurance, Travel, Telecommunications and Pro Audio.

Our purpose is to deliver superior results in the Audio market with ground breaking technology and nice user-friendly graphic interfaces.



DiBiQuadro full logo



DiBiQuadro compact logo

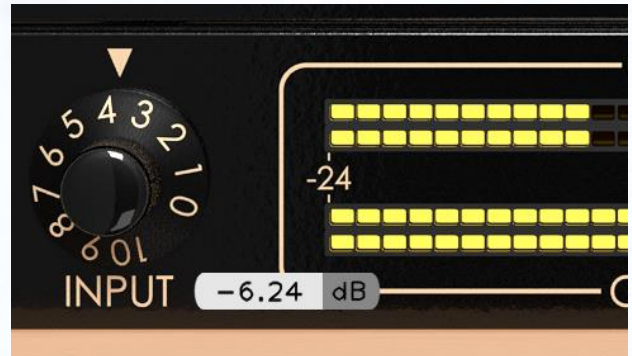
Input processing

Mercury2CP – Plugin or Renderer – has been designed to process exclusively mono sources. Therefore, a dedicated instance of Mercury2CP has to be applied to each guitar signal.

GUI: tooltips and Edit Mode

The graphic engine conception has a purpose in mind: keeping a clear indication of the status of the elements without altering their 3D rendering on the scene. Tooltips serve this purpose perfectly especially when dealing with knobs status. A typical approach would be to compress the height of the knobs trying to prevent the user to commit parallax errors. As a side effect the graphic designer would be forced to take into account this constraint by limiting the 3D depth of the scene.

Another good reason to use a tooltip-based approach is to allow the update of the knobs status by using an Edit Mode.



The tooltip



EDIT mode

When using a state-of-the-art system – see minimum requirements on DiBiQuadro website – Mercury2CP GUI is able to deliver more than 60 FPS at a reasonably low CPU usage for a super fluid knob tweaking and a hyper realistic and fast meter response.

Controls

VOLUME. This knob controls the amount of distortion or overdrive in the amplifier's preamp section.

tone-stack. The tone stack it is placed between the first and the second gain stage. It includes BASS, MID and TREBLE as you would expect in any amplifier. Anyway due to its position it affects the tone in a more subtle way compared to a tone stack placed at the end of the preamp circuit.

MASTER. This knob adjusts the overall volume of the amplifier. It controls the output level of the signal, allowing you to set the overall loudness of the amp.

LEAD DRIVE. This knob adjusts the amount of gain or overdrive in the Lead channel of the amplifier. Turning it up increases the saturation and distortion, while lowering it reduces the amount of overdrive.

LEAD MASTER. The lead master knob controls the overall volume of the Lead channel. It allows you to balance the level of the Lead channel with the Rhythm channel.

BRIGHT RHYTHM. Adds sparkle to the tone. The effect is less obvious the more the VOLUME knob is increased.

SHIFT BASS. Boosts the bass before the first gain stage.

SHIFT TREBLE. Controls the treble frequency center point. The frequency is shifted downward when engaged giving a more pronounced midrange.

DEEP. Increases the lows and the gain in the preamp.

LEAD. Activates the Lead channel. Rhythm channel is selected when deactivated.

BRIGHT LEAD. Boost the high frequencies on the Lead channel.

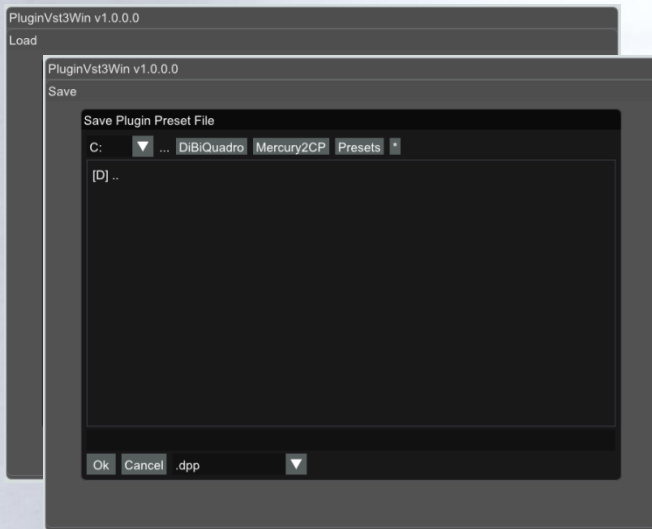
EQ. Activates the five-band EQ.

PRESENCE. The presence knob adjusts the high-frequency response.

INPUT [dB]. This knob adjusts the amount of signal entering the plugin and it can be used as a clean boost before the preamp. The INPUT LEVEL meter displays the amount of signal in input and it is directly affected by this knob.

OUTPUT [dB]. This is a clean volume knob on the general output of the plugin. Use this control to properly feed the next plugin in the chain without affecting the tone of Mercury2CP. The general output level is displayed by the OUTPUT LEVEL meter.

LOAD/SAVE. The LOAD button opens a window to select a previously saved preset – this includes knobs, buttons, switches, SIZE, TWEAK and SET settings –. The SAVE button allows to save a file containing the current settings of the plugin.



LOAD and SAVE windows

PREV/NEXT. Use these buttons to browse back and forth through the preset files.

SET/PREAMP/HEAD/POST. Please refer to the dedicated sections.

Here some useful tricks to interact with the GUI:

- use “CTRL + click” on a knob to reset the value to its default;
- use the mouse wheel on a knob for fine adjustments;
- press SHIFT while moving a knob to fine adjust.

Edit Mode

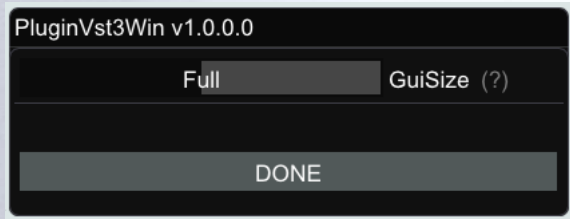
The Edit Mode is accessible by double clicking on a knob. Once done, you can edit the value by using arrows, digits and “.”, “+”, “-” keys. INSERT key is supported too: INSERT status is recognizable by checking the cursor inside the tooltip (“|” means INSERT disabled, “_” means INSERT enabled). After modifying the value press ENTER to commit or ESC to roll-back to the latest value before entering the Edit Mode.

Resize GUI

Only on Windows OS GUI is resizable by right clicking on the GUI itself and selecting one the available options. Two options are available:

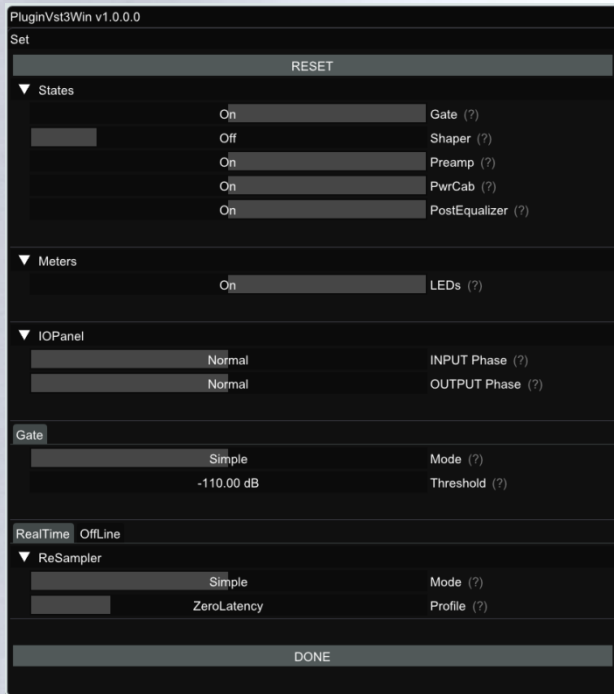
- Normal
- Full

On macOS the GUI size is automatically selected depending on the screen resolution so this feature is not necessary.



Resize GUI window (only Windows)

SET



SET window

You can open the SET window to access the configuration settings for Gate, Shaper, Preamp, Pwr/Cab,

PostEqualizer, meters, ReSampler, PreRender.

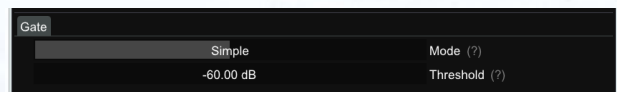
RESET. It restores the default values for all configurations in the SET window.

States. They enable / disable the various elements of the plugin.

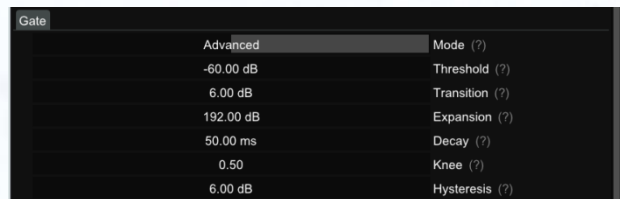
LEDs. It activates/deactivates the meter LEDs.

IOPanel. This section allows to invert the phase of the input signal (INPUT Phase) and the output signal (OUTPUT Phase).

Gate. This tab appears when the Gate is enabled in the States section. Two modes are available: Simple and Advanced.



Gate tab in Simple mode



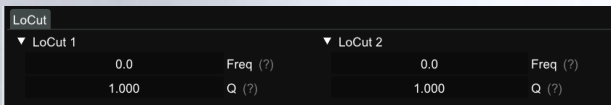
Gate tab in Advanced mode

Gate in Simple mode. A single parameter of threshold to set the level at which the gate starts to be applied. The remaining parameters are set to default values and hidden.

Gate in Advanced mode. It displays all the gate parameters. **Transition** can be used to apply a gradual gain reduction when the signal is below the threshold (0 for hard gate); **Expansion** is the maximum gain reduction when the

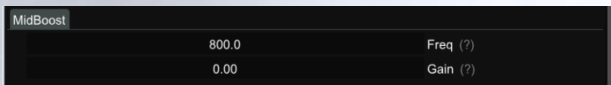
signal is below the threshold; **Decay** controls how quickly the signal fades out when it is below the threshold; **Knee** sets how hardly the gate switches from un gated to gated signal (0 for Hard Knee, 1 for Soft Knee); **Hysteresis** prevents the gate from immediately re-opening after passing below the threshold, to avoid gain flickering.

LoCut. This tab appears when the Shaper state is set to LoCut. It just contains two 12dB/oct loCut filters (Freq and Q). Particularly useful when dealing with boomy signals.



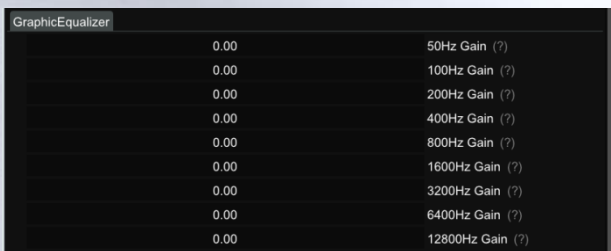
LoCut tab

MidBoost. This tab appears when the Shaper state is set to MidBoost. Use it to push the preamp on the mid frequencies.



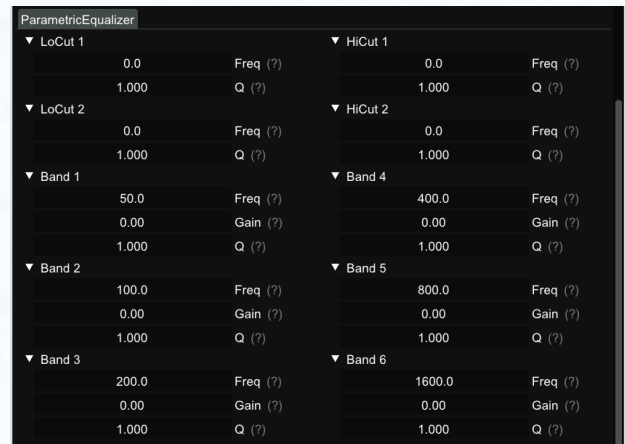
MidBoost tab

GraphicEqualizer. This tab appears when the Shaper state is set to GraphicEqualizer. Each of the nine available bands can be boosted or cut of up to 15dB.



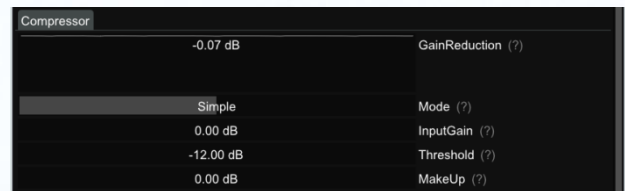
GraphicEqualizer tab

ParametricEqualizer. This tab appears when the Shaper state is set to ParametricEqualizer. If the previous shaping options were not enough, this tab offers two 12 dB/oct loCut filters (Freq and Q), two 12 dB/oct hiCut filters (Freq and Q) and 6 bell bands (Freq, Gain and Q).

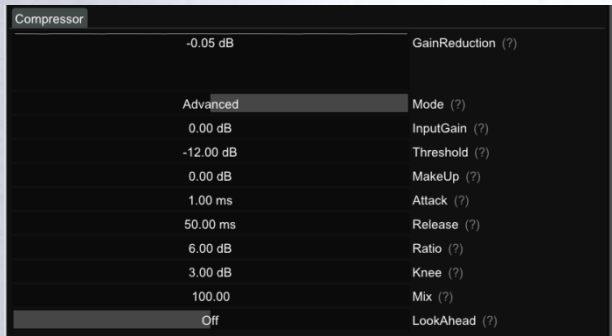


Compressor tab in Simple mode

Compressor. This tab appears when the Shaper state is set to Compressor. If you wish to add a touch of compression to your signal before hitting the preamp, this is what you are looking for. Two modes are available: Simple and Advanced. Both modes report a GainReduction indicator in dB.



Compressor tab in Simple mode



Compressor tab in Advanced mode

Compressor in **Simple** mode. Three parameters are available: **InputGain** adjusts the level of the signal before any compression happens; **Threshold** adjusts the level above which the compressor starts attenuating the input signal; **MakeUp** adjusts the level of the output signal after the compression has been applied. All the other parameters are set to default values and hidden.

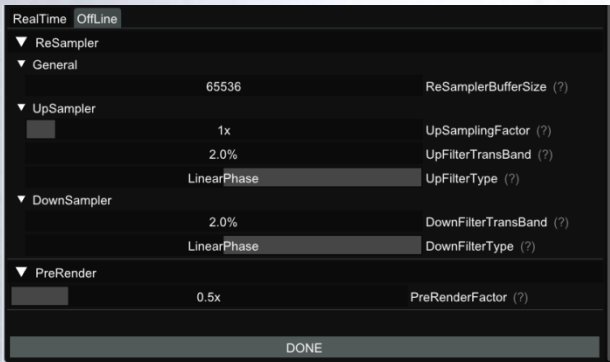
Compressor in **Advanced** mode. In addition to the parameters available in the Simple mode, the following ones are displayed: **Attack** sets the time that determines how fast the compression will set in once the signal exceeds the threshold; **Release** sets the time that determines how fast the compressor will recover from the gain reduction once the signal falls under the threshold; **Ratio** determines how much the signal is attenuated above the chosen threshold; **Knee** gives a rounder compression curve when set higher than 0 (0dB: hard knee); **Mix** balances the mix between the dry and processed signal (100%: Wet / 0%: Dry); **LookAhead** enables the compressor

LookAhead and shows the **LookAheadDelay** parameter.

RealTime and Offline tabs. These tabs allow the user to specify different configurations for the ReSampler depending on the current processing mode.

RealTime tab. This mode is applied by your sequencer when you are using a plugin for monitoring, or when you are playing tracks. We suggest to keep the UpSamplingFactor at 1x in case of monitoring, 4x in case of playback @ 44100/48000Hz, 2x in case of playback @ 88200/96000Hz.

Offline tab. This mode is set by your sequencer when you are bouncing a track or the complete mix. If the elaboration time is not a constraint for you and you need to render hi-gain guitars with the Lead channel, we suggest to set the UpSamplingFactor to at least 128x @ 44100/48000Hz, 64x @ 88200/96000Hz in order to minimize the aliasing artifacts and get super high-quality distortion. Try also to set a very high PreRenderFactor (32x) to stabilize the model in advance. **For very long renders, we strongly suggest to use Mercury2CP Renderer in command line mode.**



SET window – OffLine tab

ReSamplerBufferSize. This is the ReSampler Buffer Size in number of samples.

UpSamplingFactor. This is the UpSampling Factor applied to the base input SampleRate. Use it to increase the quality or your playbacks and renders, but take into account that high UpSampling factors correspond to more samples to be processed, and consequently longer processing times.

UpFilterTransBand. This is the UpSampling Filter Transition Band, in percent of the spectral space of the input signal between filter's -3 dB point and the Nyquist frequency.

UpFilterType. It defines the UpSampling filter's phase response: Minimum Phase or Linear Phase.

DownFilterTransBand. This is the DownSampling Filter Transition Band, in percent of the spectral space of the output signal between filter's -3 dB point and the Nyquist frequency. If you are working at sample rates higher than 44100/48000Hz, we suggest to increase it to 43% to have a smoother

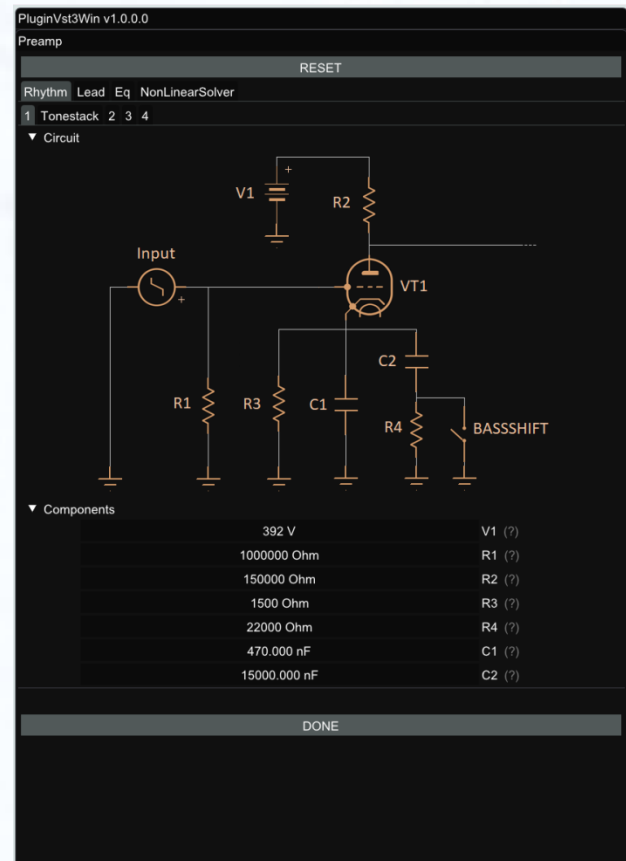
low pass filter without impacting the audible frequencies.

DownFilterType. It defines the DownSampling filter's phase response: Minimum Phase or Linear Phase.

PreRenderFactor. Together with input SampleRate and UpSamplingFactor, PreRenderFactor affects how many samples will be pre-rendered by the model before the real processing. We suggest to set it to the highest value – 32x – in Offline mode.

DONE. It closes the SET window.

PREAMP



PREAMP window

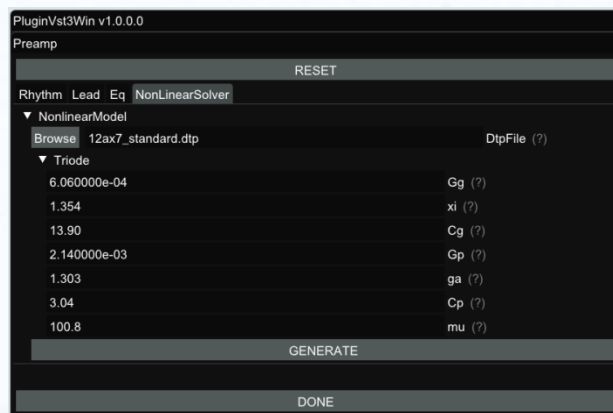
The PREAMP window gives you the power to modify every single component of the circuit. Even the Triode model of the 12AX7 can be tweaked in the NonLinearSolver tab. Play with this window very carefully, possibly in front of an analyzer, to get exactly what you are looking for. We **do not** guarantee that any possible combination of values makes sense, so it's up to you to find a configuration that works and satisfies your needs.

RESET. It restores the default values for all configurations in the PREAMP window.

Primary tabs. The primary tabs provide access to the different channels of the circuit and to the 12AX7 triode model. Each channel can be tweaked independently, while there can be only one triode model for all triodes of Mercury2CP.

Normal tab. It shows the different part of the Normal circuit, split into 4 sub-tabs, one for each gain stage, plus a Tonestack tab.

Overdrive tab. It reports a sub-tab for each of the 5 different gain stages, plus its related Tonestack.



PREAMP window – NonLinearSolver tab

NonLinearSolver tab. Use this section to change characteristics of all the triodes in the preamp. It is the most advanced and consequently the least user-friendly feature. Browse button opens the dialog window to select an existing dtp file. The dtp file contains the triode characteristic. The first time a file is selected, it is necessary to generate the look up tables (LUTs) for that triode by pressing GENERATE. Change the name in Browse text input to save a new dtp file and then press GENERATE.

The generation has to be done each time the triode parameters are modified in order to make them effective.

Once the LUTs are generated they are stored in files in the same path of the dtp file and automatically loaded when selecting the dtp file with the Browse button / dialog window.

To create a new triode characteristic follow these steps:

1. Modify the name in the DtpFile textbox.

2. Change one or more parameters of the triode model¹.

3. Press **GENERATE** button.

DONE. It closes the SET window.

If you accidentally overwrite the characteristics of the factory triodes we recommend reinstalling the product from Center.

PWR/CAB

The PWR/CAB section allows to quickly switch between 8 preloaded Pwr/Cabs. A Pwr/Cab is the non-linear representation of the chain including:

1. the power amp
2. the cabinet with the speaker
3. the microphone to capture the sound of the speaker
4. the high-end mic preamplifier

A Pwr/Cab with only the power amp is also provided: use this Pwr/Cab to experiment with "larger than life" clean sounds, or to drive an external IR (in this case an IR loader is necessary).

Mercury2CP comes with 22 selectable Pwr/Cabs:

Mercury2CP_000_75W_6L6
Mercury2CP_001_75W_6L6_60A_1_T75_57_A12C
Mercury2CP_002_75W_6L6_60A_1_T75_201_A12C
Mercury2CP_003_75W_6L6_60A_1_T75_421_A12C
Mercury2CP_004_75W_6L6_60A_4_T75_57_A12C
Mercury2CP_005_75W_6L6_60A_4_T75_201_A12C

Mercury2CP_006_75W_6L6_60A_4_T75_421_A12C
Mercury2CP_007_75W_6L6_60A_2_V30_57_A12C
Mercury2CP_008_75W_6L6_60A_2_V30_201_A12C
Mercury2CP_009_75W_6L6_60A_2_V30_421_A12C
Mercury2CP_010_75W_6L6_60A_3_V30_57_A12C_A
Mercury2CP_011_75W_6L6_60A_3_V30_201_A12C_A
Mercury2CP_012_75W_6L6_60A_3_V30_421_A12C_A
Mercury2CP_013_75W_6L6_60A_3_V30_57_A12C_B
Mercury2CP_014_75W_6L6_60A_3_V30_201_A12C_B
Mercury2CP_015_75W_6L6_60A_3_V30_421_A12C_B
Mercury2CP_025_75W_6L6_M212_V30_57_A12C_A
Mercury2CP_026_75W_6L6_M212_V30_201_A12C_A
Mercury2CP_027_75W_6L6_M212_V30_421_A12C_A
Mercury2CP_028_75W_6L6_M212_V30_57_A12C_B
Mercury2CP_029_75W_6L6_M212_V30_201_A12C_B
Mercury2CP_030_75W_6L6_M212_V30_421_A12C_B

The power amp is a recreation of a 75Watt 4x6L6 tubes with a mixed class A and class AB amplifier.

Two cabinets are available: slanted 60A loaded with T75s and V30s and M2x12 loaded with a V30 and a T75. A sequential number from 1 to 4 indicates which speaker has been sampled: 1 and 2 are the top left and right while 3 and 4 the bottom left and right respectively.

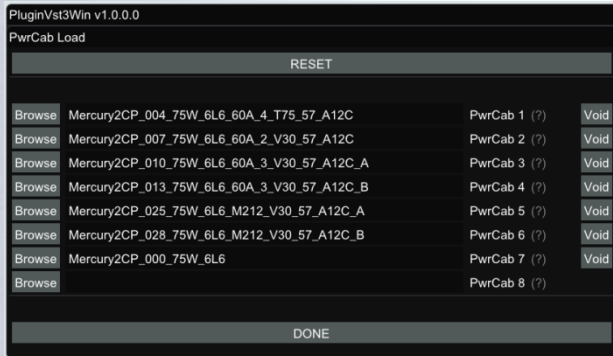
A hi-end mic preamps was used: A12C is a top notch preamplifier that set the standard for modern recordings.

All Pwr/Cabs are phase aligned (with the sole exception of the power amp Pwr/Cab).

PWR/CAB LOAD. Pwr/Cab Load gives access to the 8 slots on the GUI. Browse opens the window to select the Pwr/Cab path, while Void cleans up the slot. When a slot is void the output is taken directly from the preamp if enabled. While the output of the preamp section is not bounded by any

¹ Have a look at https://www.dafx.de/paper-archive/2011/Papers/76_e.pdf for more details.

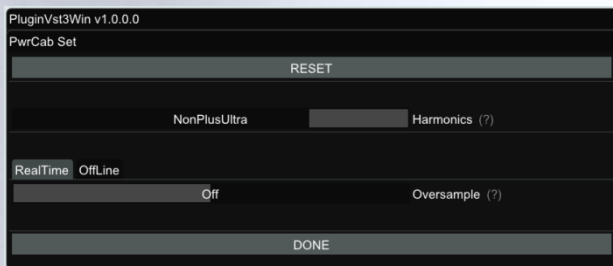
volume limit and can be used to drive subsequent plugins (ie: another Mercury2CP), the output of a Pwr/Cab is protected by a safety mechanism that clips the signal when above 0dB.



Pwr/Cab LOAD window

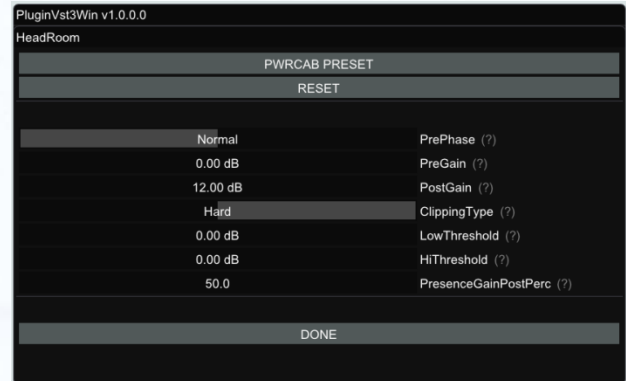
PWR/CAB SET. With Pwr/Cab Set you can select the number of harmonics in the Pwr/Cab section:

- Standard: 10 harmonics
- High: 20 harmonics
- Ultra: 30 harmonics
- Non-Plus-Ultra: 40 harmonics



Pwr/Cab SET

HEAD



HEAD window

Head (HeadRoom) is a powerful tool to shape the behavior of the power amp section including the type of clipping when pushed to the limit (soft/hard, symmetrical/asymmetrical) and the effect of depth and presence when power amp is working in the non-linear region. Each Pwr/Cab has its own setting.

PWR/CAB PRESET. It restores the factory preset for the selected Pwr/Cab.

RESET. It restores the values to the general default values.

PreGain. It sets the level of the signal in input to the power amp.

PostGain. It defines the level of the signal coming out of the Pwr/Cab.

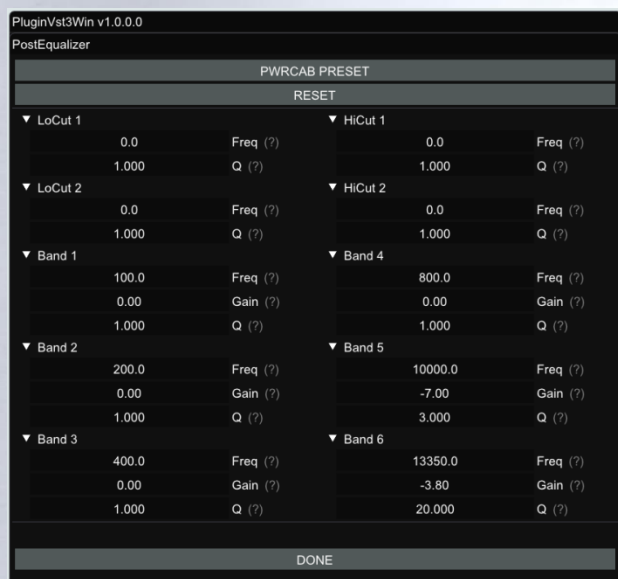
ClippingType. Soft/Hard.

LowThreshold/HiThreshold. These can be used to set the clipping thresholds of negative and positive half waves respectively: use different values for

asymmetrical clipping and identical values for symmetrical.

PresenceGainPostPerc/DepthGainPostPerc. Percentage of Presence and Depth effect after power amp clipping. They affect the behavior of Presence and Depth only when power amp is saturating.

POST



POST window

Post (PostEqualizer) is a handy minimum-phase equalizer at the end of the signal chain, with the same bands of the ParametricEqualizer available in the Shaper section.

PWR CAB PRESET. It restores the factory preset for the selected Pwr/Cab.

RESET. It restores the values to the general default values.

The Renderer



Renderer window

The Renderer is provided to process an input wave file in OffLine mode. It collects all the controls described in the previous sections, plus you have the possibility to specify **input and output files**.

Use the **LOAD** and **SAVE** buttons to load and save Renderer presets. The **Import** button can take a specific panel setting from a Mercury2CP plugin preset file. There is also an additional **Output** header which allows to specify the output wave file format.

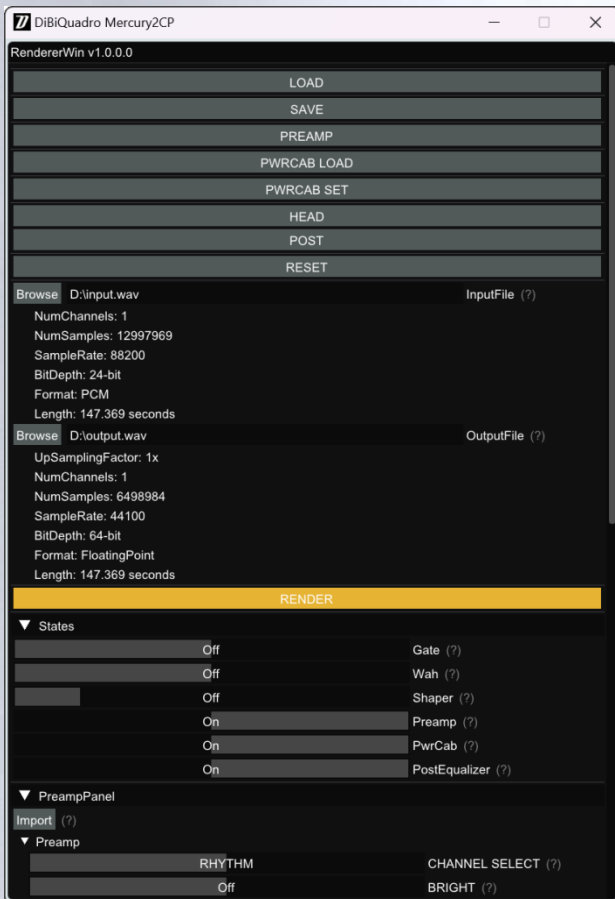


Renderer – Output header

OutputBitDepth. This is the number of bits of information in each output sample. It can be 16, 24, 32 or 64-bit.

OutputAudioFormat. It represents the output encoding type: PCM (Integer) or Floating Point.

The **RENDER** button will be **visible only** when an **existing input wave file** and a **valid output wave file name** have been specified.



Renderer – RENDER button

Command Line Mode

The Renderer can be executed in command line mode. With this functionality, multiple rendering sessions can be easily triggered in parallel by a Windows batch file.

To launch a render from the Command Prompt, first you need to create a preset for the Renderer. Then you need to execute the Renderer from the Command Prompt with “-c” option and to specify, in order:

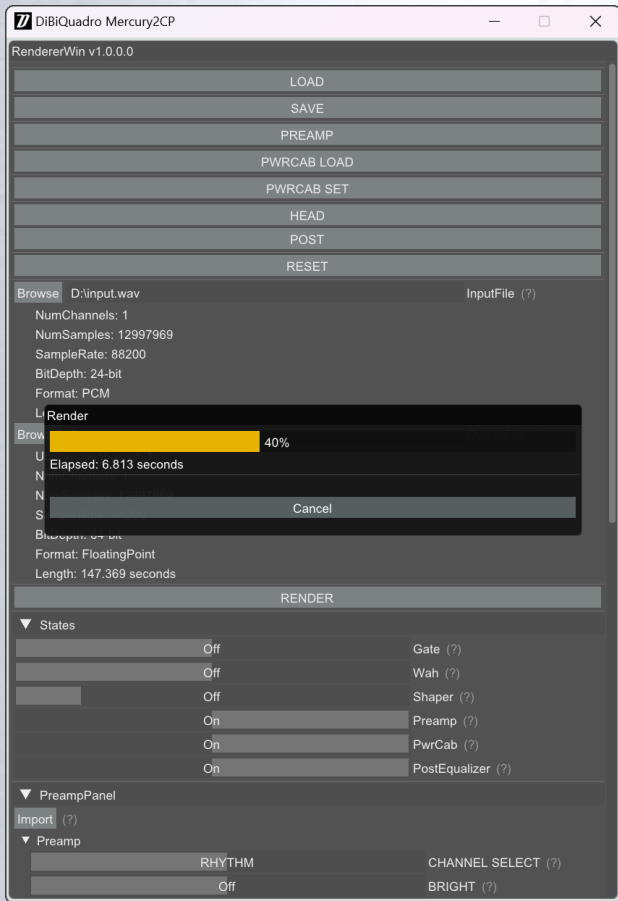
1. the full path (absolute or relative path + name) of the Renderer preset;
2. the full path (absolute or relative path + name) of the input wave file;
3. the full path (absolute or relative path + name) of the output wave file.

Reminder:

```
Mercury2CPRenderer.exe -c presetFile.drp input.wav output.wav
```

```
C:\Program Files\Common Files\VST3\DiBiQuadro\Mercury2CP>Mercury2CPRenderer.exe -c presetFile.drp input.wav output.wav
```

Example of Prompt command



Renderer – Render in progress

DiBiQuadro Audio S.r.l.s.

P.IVA: 09771000966

Address: Via Amendola 18, 26815
Massalengo (LO), ITALY

E=mc²