

PALETTE

BRUSH PACK 01 • MELODICS

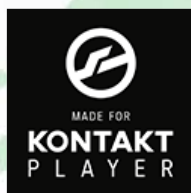
Reference Manual



RRA002 | v.1.1

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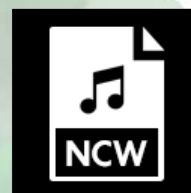
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**WORKS WITH THE
FREE KONTAKT PLAYER**



**COMPATIBLE WITH
COMPLETE KONTROL**



**LOSSLESS NCW SAMPLES
24 BIT / 44.1 KHZ**

WELCOME TO PALETTE



Palette is a series of thematic professional orchestral and cinematic Kontakt virtual instruments for songwriters and media composers. It was recorded in a dedicated hall with world-renowned performers by a passionate and experienced team. It was a dream come true for us to make and we hope this expansive set of scoring tools inspires you to write some great music.

The concept for our first Palette Brush pack **"Melodics"** is simple. We studied the scores of several of our favorite classical and modern film music composers and selected 18 of the most evocative melody-carrying combinations and instruments. We recorded a soaring violin section playing octaves, regal trumpets with horns, playful flutes & piccolo, creepy bass clarinet & contrabassoon octaves, a gigantic and downright frightening low strings & low brass ensemble and many others. We felt it was also important to include **soloists** since they are so often used to deliver a melody, so we recorded 6 of those as well.

17 of the 18 instruments in the library have **true legato**.

As with the entire Palette line, our players were recorded in standard orchestral seating position with 3 microphone perspectives – Close, Decca Tree and Hall. The room is spacious but fairly dry, which allows for easy blending with other libraries using your favorite reverbs so you're not limited to the sound of one particular hall. Since **Palette - Melodics** was recorded by the same amazing musicians and production team with the same equipment and in the same hall, it was designed to blend perfectly with all the other **Palette** libraries.

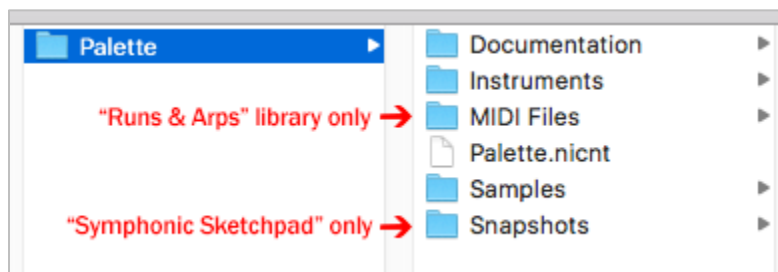


DOWNLOAD & INSTALLATION

*If this is your **FIRST** Palette product...*

1. From your User Account downloads page (or your email), download all the .rar files. You should have one .rar titled **"Palette ... Main"** and several numbered .rar's starting with **"Palette ... Samples.part01."** Check that their file sizes match what's listed in your download email.
2. To properly install the library you'll need to use .rar extraction software. PC users can download the free [WinRAR](#) application, or Mac users download the free [Keka](#) app.

3. With the proper RAR extraction app installed, **double-click the "Palette ... Main.rar."** When the extraction is complete you should have a folder titled **"Palette"** with contents that match the illustration here -> *The "MIDI Files" and "Snapshots"*



- folders are exclusive to our "Runs & Arps" and "Symphonic Sketchpad" libraries, respectively.*
4. **Double-click the "Palette ... Samples.part01.rar"** file to begin extracting your samples. You do not need to extract the other numbered .rar files as they will be extracted automatically in sequence. To ensure proper installation, make sure everything gets extracted into the same folder location as your "Palette" folder from step 3 above. If separate folders are created, you will need to combine/merge their contents into a single "Palette" folder.
 5. If you've downloaded and extracted the files correctly, you should now have one folder labeled "Palette" with contents that still match the illustration above. You can now safely move it to the hard drive location of your choice and delete all the .rar files.
 6. If you don't already have Kontakt or the free Kontakt Player, get them [here](#). Palette requires version 5.6.8 or above. If you don't already have Native Access, get it [here](#).
 7. **Run the Native Access application** and log into your Native Instruments account. **Click the "Add a serial" button**, which opens a window where you can enter the serial number you received with Palette. Copy the serial number from your download email, click inside the first box and paste it. **Click the "Add Serial" button.** On the next page, **click the "Browse" button** and navigate to the location of your library on your hard drive. Make sure to select the main Palette folder and not one of the subfolders, **then click "Open."** Check that the correct file path is displayed under the name of the library. **Click "Install"** to complete the installation process. When you launch Kontakt the library should now appear in the Libraries tab.

*If you **ALREADY OWN** other Palette products...*

1. Just repeat steps 1-4 above, making sure to extract your .rar files into your original Palette folder. All your Palette products live together in the same master "Palette" folder and use your original, single serial number. You do not need to register it a second time, so you can skip steps 6-8 above.




TIP - We strongly recommend performing a "batch resave" process on your master Palette folder after installation to shorten load times and optimize performance. For easy instructions [click here](#).

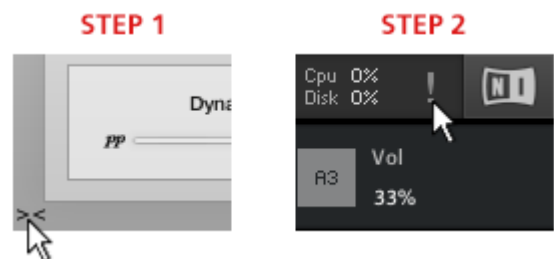
GUI

Palette - Melodics includes 18 Kontakt Instruments (.nki's). Let's go through the features of the GUI. But first, 2 very handy tips:

 ***TIP*** - We've added handy help text for every control in our GUI! If you're unclear as to the function of a control simply hover over it and the help text appears at the bottom of the Kontakt window.



 ***TIP*** - **Palette - Melodics** makes use of Kontakt 5.6.8's full available 1000-pixel wide GUI. Since some users may be viewing on laptops or lower resolution monitors, we've also included a trick to narrow the GUI to eliminate the fancy artwork and display only the primary, usable portion.



To go narrow, simply click on the small double-arrows in the lower left of the GUI. Then, to complete the transformation, click on the exclamation point (!) in the top right corner of Kontakt. Voila! Repeat this same process to go wide again.



Articulations

At the top of the GUI you'll find the Articulations panel. The list of buttons on the left displays the names of the available articulations and how each is triggered in parenthesis. Use the radial buttons to unload articulations you don't need, which helps to save RAM.

The articulation buttons have 4 possible visual states:



Clicking an articulation button selects it (represented by an asterisk) and opens its editing controls in the right side of the Articulations panel. Here you'll find **TACT**, which stands for Total Articulation Control Technology. Developed by our partners at Impact Soundworks, TACT is a powerful yet easy-to-use solution for customizing how an instrument's articulations are triggered. Starting at the top of the TACT box there are knobs for adjusting Volume, sample Offset, Attack, Release and a dropdown to select your preferred method of Dynamic Control (Velocity or MIDI CC). The red "hamburger" icon opens a menu for saving and loading your own TACT settings.

It is important to note that our legato technique crossfades between 2 samples – a sustain and a legato transition (and back again). These 2 articulations are controlled separately in TACT. So when you select and adjust the parameters of the legato articulation, only the transition samples are being affected – not the sustain. To change the parameters of your overall legato *performance*, adjust both articulations.

In the lower half of the TACT box you'll find 2 "**Rules**" that determine how the selected articulation is triggered. There are 5 options:

Key: When this is selected the articulation is triggered by pressing the note, or keyswitch, displayed in the second dropdown, which can of course be changed to your preference. Keyswitches are colored red on the Kontakt keyboard and NKS compatible controllers. The currently selected keyswitch is colored green. Note that you can activate multiple keyswitches together to stack articulations!



"Latching" means that the articulation remains active after you release the keyswitch. Deselecting this button means that the articulation will only be active when the key is held down. Note that, for obvious reasons, it is not possible to set a keyswitch in the blue playable instrument range.

Vel: When this is selected the articulation is triggered when notes are played within a designated velocity range, which you can set by dragging the values up or down in the range boxes..

CC: When this is selected the articulation is triggered by the MIDI CC you set in the first box, and only when you play within the range determined by the next two boxes.



WARNING - CC's 124-127 are forbidden as they are used internally and may cause a MIDI feedback loop that could result in a crash.

Range: When this is selected the articulation is triggered when you play a note that lies between the values you set in the two boxes.

Pedal: When this is selected the articulation is triggered when you press/activate a sustain/control pedal (also known as CC64).



TIP - Try using rules to layer 2 different articulations. This is useful when, for example, you'd like to create a sforzando by overlapping Staccato and Sustain. Set Sustain's Rule 1 to a Keyswitch and the Staccato's Rule 1 to a Velocity range, maybe 110-127. Now when you hit a Sustain hard you get sforzando!

In most cases, like a typical keyswitched setup, one Rule will be sufficient. But there may be times when you'd activate a second. For example, you might set all the short articulations to the same Rule 1 keyswitch, but then set Rule 2 to Velocity to trigger each one with a different velocity range for quicker performance (e.g. Marcato 00-40, Tenuto 41-70, Staccato 71-100, Spiccato 101-127). Another example is to set Rule 1 for both kinds of trills to the same keyswitch and set the WT Trill's Rule 2 to Pedal. There are many possibilities.

Notice that when you change an articulation's rule(s) that gets updated on the articulation button so you can always see at a glance how each is currently set to be triggered.

Microphones

The second panel of the GUI contains the microphone controls. For each of the 3 mic positions - **Close, Decca and Hall** - there's an on/off button (turn off unneeded mics to save RAM), volume fader, pan knob, mute & solo buttons and a button to select the output channel for cases when you'd like each mic routed to different tracks in your DAW. Patches that feature more than one instrument, or instruments playing in octaves, have an **additional Close mic** position. In these cases each is generally focused on one of the two instruments in the combo so you can subtly adjust the mix. The red "hamburger" button allows you to save/load mixer settings.

Settings

The third (center) panel is where you can adjust various performance settings.

Round Robin: Turning this off deactivates the additional round robin samples, which makes the instrument sound more robotic. It is recommended that this be left on for more realistic results. The switch changes the order of playback of the round robins from Cycle to Random.

Releases: Releases provide a more natural, realistic ending to sustained notes when you release the key. Again we recommend these be left on for a best results. You can use the slider to adjust their volume.

HQ Vel Xfade: When this is active Kontakt employs AET filters on all short articulations to smooth out the steps between dynamics layers for more natural results. Note that this does, however, use more CPU.

Velocity: These 3 knobs allow you to customize the velocity curve for additional control over dynamics. This will be most useful for short articulations and when Dyn Control is set to Velocity in TACT.

FX

The fourth panel houses a compact yet powerful 8-slot FX rack. To load an effect, click on any slot's down-facing arrow. This opens a dropdown where you can select one of 17 effects, or move them up or down in the rack to change their order. The radio buttons turn the slot on and off (bypass).

Slots have 3 states – **empty**, **loaded** and **selected**.



Clear (white) signifies that the slot is **empty** (no effect currently loaded).



Gray means there's an effect **loaded** and its name is displayed, but it's not selected for editing.



Red means there's an effect loaded and it's **selected** for editing. Its controls appear on the right.

The red "hamburger" icon opens a menu for saving and loading tweaked versions of individual effects as well as your entire FX rack. Full details for all 17 effects can be found in the **FX** part of this manual.

Performance Sliders

The fifth and final panel on the GUI contains sliders for Dynamics, Expression (volume) and instrument-specific articulations like Vibrato, Tremolo and Flutter (where applicable). Right-click on a slider to MIDI-learn (assign) it to the CC of your choice. Here are the default settings for the entire library:

Dynamics: CC1

Expression: CC11

Vibrato: CC4



****WARNING* - CC's 124-127 are forbidden as they are used internally and may cause a MIDI feedback loop that could result in a crash.***

LIST OF ARTICULATIONS

VIOLINS OCTAVES (6-5)

True Legato
Sustain
Sustain Vibrato
Marcato
Tenuto
Staccato
Spiccato
Major & Minor Trills

VIOLINS OCTAVES (CON SORD) (6-5)

True Legato
Sustain
Sustain Vibrato
Marcato
Tenuto
Staccato

CELLI + DOUBLE BASSES OCTAVES (6-4)

True Legato
Sustain
Sustain Vibrato
Marcato
Tenuto
Staccato
Spiccato

FLUTES + PICCOLO OCTAVES (2-1)

True Legato
Sustain
Sustain Vibrato
Marcato
Tenuto
Staccato
Staccatissimo
Rips

OBOE + BASSOON OCTAVES (1-1)

True Legato
Sustain
Sustain Vibrato
Marcato
Tenuto
Staccato
Staccatissimo
Rips

BASS CLARINET + CONTRABASSOON OCTAVES (1-1)

True Legato
Sustain
Marcato
Tenuto
Staccato
Rips

TRUMPETS + HORNS OCTAVES (3-3)

True Legato
Sustain
Sustain Vibrato
Marcato
Tenuto
Staccato
Staccatissimo

TROMBONES + TUBA (2-1-1)

Sustain
Marcato
Tenuto
Staccato

6 HORNS (6)

True Legato
Sustain
Marcato
Tenuto
Staccato
Staccatissimo

TRUMPETS OCTAVES (2)

True Legato
Sustain
Marcato
Tenuto
Staccato
Staccatissimo

LOW BRASS + LOW STRINGS (2-1-1 + 6-4)

True Legato
Sustain
Marcato
Tenuto
Staccato
Staccatissimo

HORNS + VIOLINS OCTAVES (3 + 6-5)

True Legato
Sustain
Marcato
Tenuto
Staccato
Staccatissimo

SOLO VIOLIN

True Legato
Sustain
Sustain Vibrato
Marcato
Tenuto
Staccato
Spiccato
Pizzicato
Tremolo
Measured Tremolo
Major & Minor Trills

SOLO CELLO

True Legato
Sustain
Sustain Vibrato
Marcato
Tenuto
Staccato
Spiccato
Pizzicato
Tremolo
Measured Tremolo
Major & Minor Trills

SOLO FLUTE

True Legato
Sustain
Sustain Vibrato
Marcato
Tenuto
Staccato
Staccatissimo
Major & Minor Trills
Rip
Double & Triple Tongue

SOLO OBOE

True Legato
Sustain
Sustain Vibrato
Marcato
Tenuto
Staccato
Staccatissimo
Major & Minor Trills
Double & Triple Tongue

SOLO TRUMPET

True Legato
Sustain
Sustain Vibrato
Marcato
Tenuto
Staccato
Staccatissimo
Major & Minor Trills
Rip
Fall
Double & Triple Tongue

SOLO HORN

True Legato
Sustain
Sustain Vibrato
Marcato
Tenuto
Staccato
Staccatissimo
Major & Minor Trills
Rip
Fall
Double & Triple Tongue

Palette - Melodics features an 8-slot FX rack with 17 high quality Kontakt FX. Here are definitions and an explanation of controls for each available effect.



EQ

The Solid G-EQ is modelled on high quality analogue circuitry. It is a 4-band parametric EQ and offers the choice of bell or shelf style control of the low and high frequency bands.

Controls

Low, Low Mid, High Mid, High: Adjusts the amount of boost or cut at each frequency.

Freq: Adjusts the center frequency of the frequency band at which the boost or cut will occur.



Solid Bus Compressor

The Solid Bus Compressor is modelled after a classic analog bus compressor. It offers a more characteristic dynamic control than Kontakt's standard compressor.

Controls

Threshold: Sets a level threshold above which the Compressor starts working. Only levels that rise above this threshold will be reduced by the compression; signals that stay below it will be left unprocessed.

Ratio: Controls the amount of compression, expressed as a ratio of "input level change" against "output level change". A ratio of 1:1 means that no compression will be happening. For example, a setting of 4 represents the

ratio 4:1, which means for every 4 decibel increase of amplitude above the threshold, the output will increase by only 1 decibel.

Attack: Adjusts the time the compressor will take to reach the full Ratio value after an input signal exceeds the Threshold level.

Release: Adjusts the time the compressor will take to fall back to non-compression after the input signal falls below the threshold.

Makeup: Controls the output gain of the compressed signal. Used to compensate for the gain reduction of the effect.

Mix: Controls the dry/wet mix of the compressor. This can be used to create a parallel compression style routing, which increases the quieter signals rather than reducing the louder ones. At a setting of 100% you will only hear the compressed signal, at a setting of 0% you will only hear the unprocessed input signal.

Output: Controls the module's output level.



Feedback Compressor

The Feedback Compressor is a type of compressor that compares the amplifier's output signal, rather than the input signal, to a threshold level. When the threshold level is reached, the compressor reduces the signal's gain level. The Feedback Compressor is modeled after a classic feedback compressor known for its bright and punchy sound.

Controls

Input: Adjusts both the input level and the threshold simultaneously. Turning this knob clockwise will result in more compression.

Ratio: Controls the amount of compression, expressed as a ratio of "input level change" against "output level change". A ratio of 1:1 means that no

compression will be happening. For example, a setting of 4 represents the ratio 4:1, which means for every 4 decibel increase of amplitude above the threshold, the output will increase by only 1 decibel.

Attack: Controls the scaling of the attack phase of the input signal's volume envelope. Increasing this parameter will add more punch and decreasing it will reduce sharp attacks.

Release: Determines how long it takes for the compression action to stop after the input signal falls below the threshold level. Typical values range from 50 to 250 ms.

Makeup: Controls the output gain of the compressed signal. Used to compensate for the gain reduction of the effect.

Mix: Controls the dry/wet mix of the compressor. This can be used to create a parallel compression style routing, which increases the quieter signals rather than reducing the louder ones. At a setting of 100% you will only hear the compressed signal; at a setting of 0% you will only hear the unprocessed input signal.

Output: Controls the module's output level.



Transient Master

The Transient Master is an easy to use compressor designed to control the attack and sustain of a sound. Instead of following the amplitude of the sound like a traditional compressor, it follows the general envelope and is thus not as susceptible to changes of input gain. It is best used on sounds with fast attacks, like percussion, pianos or guitars. The Transient Master can also be quite extreme in some cases, so use it with caution.

Controls

Input: Controls the input gain to the effect.

Attack: Controls the scaling of the attack portion of the input signal's volume envelope. Increasing this parameter will add more punch and decreasing it will reduce sharp attacks.

Sustain: Controls the scaling of the sustain portion of the input signal's volume envelope. Increasing this parameter will add more body to the sound and decreasing it will reduce the sound's tail.

Output: Controls the output gain after the effect.



Limiter

The Limiter acts as a "safety net" to keep short signal peaks from overloading the system, which would result in audio clipping. While compressors have a range of artistic applications, limiters are usually used for technical reasons - they can tame signals with peaks which would otherwise overload the output, without requiring you to turn the signal's overall volume down.

Controls

Input: Sets the gain of the input signal. The Limiter is different from the Compressor in that it has a fixed threshold; to achieve a sensible peak reduction, use this control to adjust the input gain until you see the Attenuation meter responding only to occasional level peaks.

Release: Just like the Compressor's control of the same name, this knob adjusts the time it takes the Limiter to return to an unprocessed signal after the input level falls below the threshold.

Output: Adjusts the module's output level.



Tape Saturator

The Tape Saturator emulates the soft compression and distortion of recording to tape. It can be used lightly to add warmth and coloring to the sound, or heavily to add aggressive distortion.

Controls

Gain: Controls the input gain of the effect. This will increase the amount of tape distortion and compression.

Warmth: Controls the low frequency boost/cut of the effect.

Rolloff: Controls the high frequency rolloff starting frequency. Frequencies above this point will be attenuated.

Output: Controls the output gain of the effect.



Saturation

The Saturation effect is basically an amplifier with a non-linear characteristic.

Controls

Amount: Adjusts the transfer curve. A negative setting results in a characteristic that will expand the signal - lower sample values will be attenuated; higher values will be amplified. Positive settings do the opposite and thusly simulate the compression-like saturation of an analogue circuit. At a value of 0.0, the signal will pass the module unprocessed.

Output: Adjusts the module's output level.



Distortion

Distortion clips or rounds off high sample values. It thereby simulates the behavior of overloaded transistor or tube circuits, adding artificial harmonics to a sound.

Controls

Drive: Adjusts the amount of distortion.

Damping: Turning this knob clockwise attenuates high frequencies in the output signal, thereby counteracting the brightness caused by the artificial harmonics.

Output: Adjusts the module's output level. Since distortion boosts the gain considerably, it's often necessary to attenuate the signal at the output stage.



Lo-Fi

Lo-Fi adds various digital artifacts, like quantization noise or aliasing, to a clean signal. It's great for roughing up sounds that would otherwise be too plain and featureless.

Controls

Bits: Re-quantizes the signal to an adjustable bit depth. Fractional bit levels are possible and can add considerable "grit". Audio CDs have a quantization depth of 16 bits, old samplers frequently used 8 or 12 bits.

Rate (Sample Rate): Re-samples the signal to an adjustable sample rate. The re-sampling is done without any kind of (usually mandatory) low-pass filtering, which causes all kinds of wonderful aliasing artifacts. The sample

rate goes all the way down to 50 Hz, which will not leave much of the original signal.

Noise: Adds hiss to the audio signal.

Color: Adjusts the frequency characteristic of the noise and acts as a low-pass filter.

Output: Adjusts the module's output level.



Screamer

The Screamer offers an alternate overdrive algorithm that sounds warmer and smoother than the Distortion effect.

Controls

Drive: Adjusts the amount of distortion.

Tone: Controls the brightness of the sound. Turning this knob clockwise will result in a more pronounced top end. Turning it counter-clockwise results in a mellower, darker sound.

Bass: Adjusts the low frequency gain.

Treble: Adjusts the high frequency gain.

Clean: Blends clean signal into the distorted tone. At 0.0 %, only the distorted signal is audible, while at 100.0 %, equal amounts of distorted and clean signal are mixed.

Output: Adjusts the module's output level.



Amp

The Amp effect simulates the classic tone of British guitar amplifiers.

Controls

Hi Gain: Increases the preamp's gain potential. Switch to Hi Gain mode if you want to create distinctly distorted or saturated sounds.

Gain: Sets the amount of gain added by the preamp. Turning it clockwise adds drive, distortion and edge to the sound.

Bass: Adjusts the low frequency response.

Mid: Adjusts the midrange frequency response.

Treble: Adjusts the high frequency response.

Presence: Boosts the frequency response in the upper midrange.

Master: Adjusts the module's output level.



Cabinet

The Cabinet effect simulates the sound of a guitar cabinet recorded through a microphone. By following a distortion effect (like the Screamer) with this module in your insert chain, you can simulate a complete guitar amp.

Controls

Size: Adjusts the size of the simulated cabinet. Larger cabinets tend to have a more pronounced bass response, while smaller cabinets can sound thin and tinny.

Air: Controls the level of early reflections in the room response, adding a sense of space.

Bass: Boosts or cuts the level of the lower frequencies.

Treble: Boosts or cuts the level of the higher frequencies.

Type: Allows you to select the simulated cabinet model.

Output: Adjusts the module's output level.



Stereo Widener

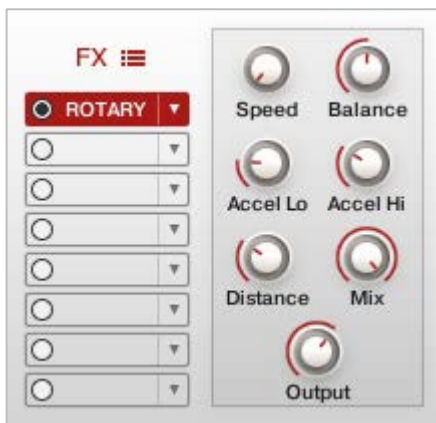
The Stereo Widener allows you to control the width of your signal's stereo base, change the panning, and create a pseudo-stereo signal from mono sources.

Controls

Spread: Collapses (counter-clockwise) or expands (clockwise) your signal's stereo base. At the far left position, stereo signals will be summed to mono. Positive values will result in an artificial widening of stereo sources that has a tendency to expand beyond the speakers.

Pan: This control allows you to place your signal within the stereo field. It works exactly like the Pan control of the Amp module.

Output: Adjusts the module's output level.



Rotary

The Rotator realistically simulates the sound of rotating speaker cabinets, which are commonly associated with drawbar organs that became popular in rock music of the 60s and 70s. Although the effect is almost intrinsically tied to "the" prototypical drawbar organ sound, it works equally well on guitars, synth pads, and a wide range of other sounds.

Controls

Speed: Although this parameter appears as a knob in order to facilitate automating, it really only has 2 positions - Slow and Fast. A change of this setting realistically simulates the acceleration or braking of the rotor.

Acceleration and Brake Speed: These adjust how quickly the rotors of the treble (Accel Hi) and bass (Accel Lo). At the rightmost position, the respective speaker will change its speed instantly, while it will take a long time to reach its designated speed with the fader at the leftmost position.

Balance: Controls the relative levels of the cabinet's treble and bass parts.

Distance: Controls the simulated distance between the cabinet and the pickup microphones. A closer distance results in a wider stereo panorama.



Chorus

The Chorus module "thickens" the audio signal by splitting it up and detuning one version in relation to the original. Separate LFOs with an adjustable phase relationship detune each stereo channel independently for creating wide-panorama effects.

Controls

Speed: Adjusts the LFO speed.

Depth: Adjusts the range of modulated detuning. Higher values give a more pronounced chorusing effect.

Phase: Imparts an LFO phase difference between the left and the right stereo channel. This can considerably increase the width of the output signal's

stereo base.

Mix: Adjusts the respective levels of the original and processed signals.



Flanger

The Flanger splits the audio signal up and delays one version in relation to the original signal. By modulating the delay time, as well as feeding an adjustable amount of the output signal back into the input, the Flanger creates a characteristic "whooshing" sound. The Flanger uses a separate LFO for each stereo channel, with the phase relationship between both LFOs being adjustable.

Controls

Speed: The LFO speed.

Depth: The amount of LFO modulation. Higher values cause the flanging effect to sweep over a wider range.

Phase: Imparts an LFO phase difference between the left and the right stereo channel. This can considerably increase the width of the output signal's stereo base.

Feedback: Feeds a certain amount of the delayed signal back into the module's input, thereby creating a more pronounced effect.

Color: Adjusts the delay line's range of operation and, consequently, the color of the flanging effect. Small values result in short modulated delay times, making the Flanger sound more like a phaser.

Mix: Adjusts the respective levels of the original and processed signals.



Phaser

This effect continually changes the phase relationships in your signal with an all-pass filter. This results in a comb filtering effect, which attenuates some frequencies while boosting others. The sound is similar to that of a flanger, but in a more subtle manner.

Controls

Speed: The LFO speed.

Depth: The amount of LFO modulation. Higher values cause the phaser effect to sweep over a wider range.

Phase: Imparts an LFO phase difference between the left and the right stereo channel. This can considerably increase the width of the output signal's

stereo base.

Feedback: This control adjusts the emphasis of the peaks and notches that the comb filter effect imparts on the signal.

Mix: Adjusts the respective levels of the original and processed signals.



Delay

This effect offers a delay line that can optionally be synced to the tempo and provides an adjustable feedback level, a low-pass filter and a pan control for ping-pong echo effects. If you don't use the tempo syncing feature, the available delay range is 5 to 2900 ms.

Controls

Time: The delay time in note increments (when the SYNC button is activated) or milliseconds (when the SYNC button is deactivated).

Damping: Attenuates high frequencies in the delayed signal. Turning this control clockwise will increase the damping effect. If you have set a feedback level, the signal will gradually lose more high frequency content with each

repetition.

Pan: Setting a value higher than 0 creates a ping-pong effect. Higher values will result in wider panning; at 100, signals alternate between the far left and far right channel.

Feedback: Controls the amount of the output signal that's being fed back into the input of the delay line, thereby creating a series of echoes that gradually fade into silence.

Mix: Adjusts the respective levels of the original and processed signals. In common scenarios, the delayed signal is mixed in at a lower level than the direct signal.



Convolution Reverb

Convolution is a method for achieving highly realistic reverbs. Convolution is a sophisticated mathematical process that replicates the acoustical behavior of a room or other physical space for use with your own signals. To accomplish this, a short audio recording of a wide-band signal played through the system is fed into the convolution processor. This recording is called an impulse response (IR). Palette ships with 11 custom IR's created specifically to provide a variety of realistic performance settings for the orchestra.

Controls

Predelay: Introduces a short delay between the direct signal and the reverb trail build-up. This corresponds to the natural reverberation behavior of large rooms, where a short time elapses before the first reflection of a sound wave returns from a wall.

Size: Adjusts the size of the simulated room. This affects the duration of the reverb trail.

Tone: This control allows you to adjust the construction material of the simulated room and, consequently, the color of the reverb trail. Low values simulate softer surfaces like wood, while high values simulate the reflection behavior of hard surfaces like concrete.

Mix: Adjusts the respective levels of the original and processed signals. In common scenarios, the reverb signal is mixed in at a lower level than the direct signal.

IR: Allows you to select from 11 different simulated acoustic spaces including concert halls, churches and performance rooms.

CREDITS

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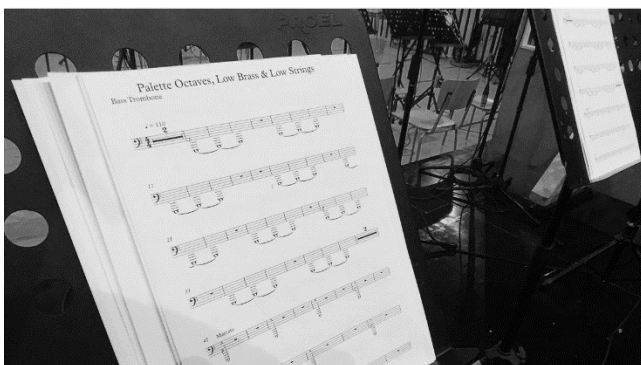
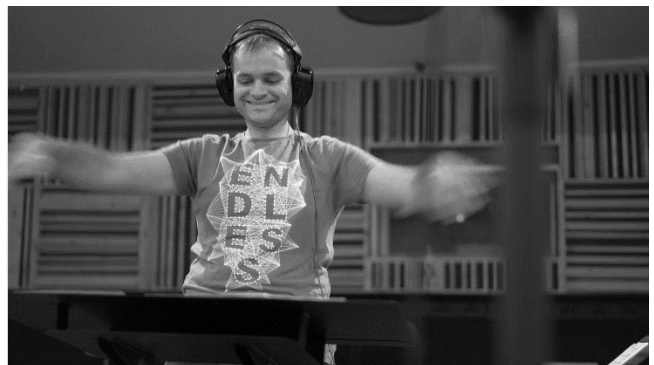
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Thanks: In addition to everyone above who worked so hard to create this library we'd also like to give extra special thanks to Jodi & Phoebe & Lola Chapin, Marcy & Wren Yoder, Jelena & Viktor Krušelj, Impact Soundworks, and the entire staff at Sofia Sessions Studios and FourForMusic.

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