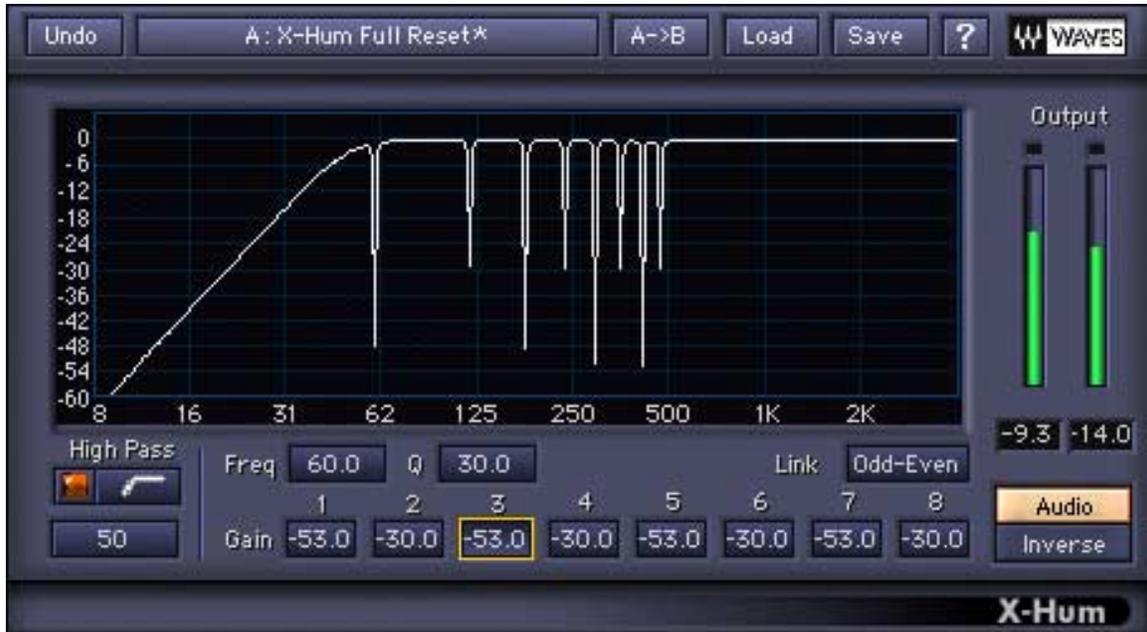


# Waves X-Hum

## Software Audio Processor

### Users Guide



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# 1.Introduction

Thank you for choosing Waves! In order to get the most out of your new Waves plugin, please take a moment to read this user guide. To install software and manage your licenses, you need to have a free Waves account. Sign up at [www.waves.com](http://www.waves.com). With a Waves account you can keep track of your products, renew your Waves Update Plan, participate in bonus programs, and keep up to date with important information.

We suggest that you become familiar with the Waves Support pages: [www.waves.com/support](http://www.waves.com/support). There are technical articles about installation, troubleshooting, specifications, and more. Plus, you'll find company contact information and Waves Support news.

**Waves X-Hum** reduces rumble, DC-offset, and hum while preserving excellent audio quality. **X-Hum** is part of the Waves **Restoration** bundle, which restores vinyl records and damaged recordings. X-Hum and the other Restoration plug-ins have a friendly interface that is easy to learn and use.

This user guide describes:

- the problems **X-Hum** solves;
- how to use the software;
- the software user interface.

## WHAT PROBLEMS DOES X-HUM FIX?

**X-Hum** effectively reduces all three of these problems:

- *Hum disturbance* is usually caused by problematic ground-loop circuits. Steady, low-frequency oscillation may occur in the recording, usually at a sub frequency of the AC used in that country. For example, Europe uses 240 VAC so an ungrounded loop can cause a 60 Hz hum. The fundamental disturbance can also cause harmonics at high enough amplitudes to create additional problems.
- *Low-frequency rumble* is caused by mechanical analog systems, such as turntables and tape machines; it is not steady in pitch.
- *DC offset* tilts in the audio waveform to one side of the zero baseline.

## HOW DOES X-HUM WORK?

Hum, rumble, and DC-offset disturbances are usually stable throughout the afflicted recording. For this reason, an EQ device is better suited for reducing the problem than a dynamic process. **X-Hum** uses higher order filters with extremely narrow cut notches than those found in a typical EQ. **X-Hum's** notches can cut up to 60 dB at very narrow bandwidths.

## 2. Using X-Hum

**X-Hum** is divided into two sections:

- A high-pass filter eliminates rumble and DC-offset.
- Eight notch filters linked in a harmonic frequency structure eliminate steady-pitched hum.

Two parameters affect the high-pass filter:

- **Slope** can be set to  $-12$  or  $-24$  dB/octave.
- **Frequency** determines the filter cutoff frequency. Use a low cutoff frequency (i.e., 10 Hz) to eliminate DC offset and preserve the signal's musically significant low-frequency content. Use a higher cutoff (i.e., 40–80 Hz) to eliminate rumble.

The *Harmonic Notch Filters* section removes steady-pitched hum with different harmonic structures built above the fundamental oscillation. There are three parameters:

- The **Frequency** control sets the fundamental center frequency of the filter.
- The **Global Q** sets the width of the notch filters. For very steady-pitched hum, use a narrow Q. If the hum's frequency shifts throughout the recording, use a wider Q.
- The **Cut Gain** of the notch filter can be set separately for each harmonic filter.

Three link modes allow different methods of altering the gain of the filters:

- **Linked**: All filters are linked such that altering one filter adjusts all filters at once while preserving their relative offsets.
- **Odd/Even**: Links the gain of filters 1,3,5,7, and 2,4,6,8 while preserving the relative offsets within the odd and even groups.
- **Unlinked**: Filters are not linked; all filters can be adjusted independently.

**NOTE:** Relative offsets are preserved until a maximum or minimum value is reached. Once a filter has reached its limit, further movement of the other filters requires unlinking.

Although some phase distortion occurs in most digital and all analog EQs, these devices do not use the extreme slopes and cuts found in X-Hum. Since phase distortion increases with these more extreme settings, we recommend using X-Hum as moderately as possible to satisfy your requirements.

In many cases, a factory preset will solve your recording's problems. If a factory preset is not suitable, find the preset that works best and alter its parameters to match the specific task.

### 3. X-Hum Controls and Displays

#### CONTROLS

##### *High-Pass*



##### **ON/OFF SWITCH:**

Turns the High-Pass filter *on* or *off*.

Default = Off

##### **SLOPE:**

Determines the order of the filter.

**Settings:** Moderate (12 dB/octave), Steep (24 dB/octave); Default = Moderate

##### **FREQUENCY:**

Sets the High-Pass filter's cutoff frequency.

**Settings:** 4–100 Hz; Default = 20 Hz

##### *Harmonic Notch Filters*



##### **FREQUENCY:**

Sets the fundamental frequency of the first filter for hum removal. The subsequent seven filters are set harmonically relative to this filter. For example a fundamental at 60 Hz would have its harmonics at:

60\*2=120 Hz, 60\*3=180 Hz, 60\*4=240 Hz, 60\*5=300 Hz, 60\*6=360 Hz, 60\*7=420 Hz.

**Settings:** 20–500 Hz; Default = 60 Hz

##### **Q:**

Sets the bandwidth of the notch filter. These notch filters are capable of very narrow Qs. Higher numbers correspond to a narrower Q.

**Settings:** 3–60; Default = 30

#### **GAIN:**

X-Hum has separate gains for each of the eight harmonic notch filters. Since these are cut filters, gains are negative so the absolute value is shown. The gains can be altered from the graph by clicking and dragging a notch's + marker. The Gain behavior resulting from dragging follows the Link mode setting (see below).

**Settings:** 0–60 dB; Default = 0

#### **LINK MODE:**

The Link mode selector has three settings that define how altering one filter's gain affects the others:

**Linked:** All filters are linked such that altering one filter adjusts all filters at once while preserving their relative offsets.

**Odd/Even:** Links the gain of filters 1,3,5,7, and 2,4,6,8 while preserving the relative offsets of odd and even groups.

**Unlinked:** Filters are not linked; all filters can be adjusted independently.

Offsets are preserved in linked mode only when gain adjustments yield settings within maximum and minimum boundaries.

#### **MONITOR**



The Monitor control toggles between Audio and Inverse.

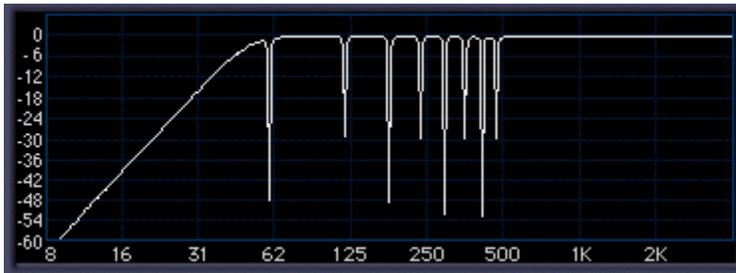
- The **Audio** path plays the processed audio; this is the normal mode to monitor X-Hum's effect.
- **Inverse** mode keeps the filters in the same position but boosts (instead of cuts) while applying appropriate gain reduction. This technique is well known to sound engineers as they search for a problematic feedback or resonant frequency while equalizing a concert environment. It is sometimes easier to spot the problem by boosting suspect frequencies to exacerbate the problem than by cutting to solve it directly. The Inverse function does not yield a boost filter shape that is perfectly symmetric to the cut filter's. The boost filters are not as narrow and never boost 60 dB. The High-Pass filter is not included in Inverse mode and is turned off.

The following figure shows the **X-Hum** graph in Inverse mode:



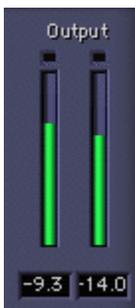
## DISPLAYS

### X-HUM GRAPH



The graph represents the **X-Hum** notch filter settings. The x-axis shows frequencies in the range 10 Hz – 4 kHz (logarithmic scale). The y-axis shows amplitude from +6 dB to –60 dB.

### OUTPUT METERS AND CLIP LIGHTS



The output meters show the output level in dBFS (dB below Full-Scale digital). The clip light above the meters lights when the output exceeds 0 dBFS. You might be wondering: How can a cut-only device produce clipping? Since X-Hum's EQ filters are not phase linear, they introduce some phase distortion of the input signal. This is common in all non-phase-linear EQs. Due to the shifting phase of some frequencies in relation to others, parts of the signal that were close to the full scale may suddenly exceed it. It is important to know when this

occurs so clipping is not mistaken for other noise. To eliminate clipping reduce the gain of the input signal.

## **WaveSystem Toolbar**

Use the bar at the top of the plugin to save and load presets, compare settings, undo and redo steps, and resize the plugin. To learn more, click the icon at the upper-right corner of the window and open the WaveSystem Guide.