

DAC & High Power Digital Preamplifier & HEADPHONE amplifier

OPERATING INSTRUCTIONS



CanEVER AUDIO®

FRONT and REAR view









Press the **SETUP** button for at least 2 seconds and LESS than 10 seconds to enter SETUP Mode.

The value of each parameter is visible on the display.

Press the SETUP button again to skip to the next parameter.

Rotate the right volume knob to change the parameter values.

If no button is pressed, or the right knob is not turned within 10 seconds, the *ZeroUno* _{DAC} automatically stores the values shown at the display and switches back to the PLAY/MUTE mode.

All selected parameters are stored in a non-volatile memory, to keep the setup information after the unit is switched off.

INFRARED REMOTE CONTROL



- Keep the INPUT button pressed and press the SETUP button again
- Release both buttons (INPUT & SETUP) and the RC is paired.

If there is still interference with other RC's in the household, please restart the process above described to generate another code.

NOTE: Each time the RC communicates with the ZeroUno _{DAC}, a dot appears at the lower-right corner of the display. If this dot does not appear, check the battery inside the RC (battery type: CR2032B)

To get instant feedback about the ZeroUno _{DAC} status, every time a button is pressed at the RC, the display shows the value in large digits for 5 seconds.

steps of 0.5dB.



steps of 1.0dB. The range is -60dB to 0dB. If the key is pressed constantly, the attenuation changes quickly.

When the VOLUME buttons are pressed, the attenuation changes in

When the **BALANCE** buttons are pressed, the balance changes in

The range is LEFT -5.0dB to RIGHT +5.0dB.

The first row of the display reports a bar showing the position of the balance value in the range of -5.0dB to 5.0dB.

When the balance is set to 0.0dB the display clearly shows the condition.



When the **MUTE** button is pressed, the *ZeroUno* _{DAC} is muted and the display *never* switches back to the standard size view until the MUTE key is pressed again.

After the MUTE button is pressed again, the *ZeroUno* _{DAC} is un-muted. The display returns to normal operation.



When the **INPUT** button is pressed, the *ZeroUno* $_{DAC}$ switches between the INPUTs.

In the last row of the display the selected INPUT is shown.



RESTORE OF FACTORY SETTINGS



The *ZeroUno* _{DAC} is completely configured by the factory for top performance.

To **RESTORE** the factory setup, press the SETUP button at the front panel of the *ZeroUno* $_{DAC}$ for at least 10 seconds.

The ZeroUno $_{DAC}$ will RESTORE, including set parameter when the SETUP button is released.

A countdown will appear on the second row of the display, when the *ZeroUno* $_{DAC}$ has engaged the RESTORE command.

Once engaged, it is not possible to stop the RESTORE command.

PARAMETERs SETUP

To enter the PARAMETERs SETUP mode, press the SETUP button for at least 2 seconds, but less than 10 seconds.

Press the SETUP button again to skip from one parameter to the next.



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STATUS LED

RED non MQA (Master Quality Authenticated) track is playing BLUE MQA studio track is playing

GREEN MQA track is playing

Turn the volume knob to choose, if you want to turn on the STATUS LED at all.

After 10 seconds without interaction at the right knob, the display returns to normal operation and the parameter is stored. The countdown on the first row helps to control the time left.

RENAME of INPUTS

Turn the volume knob to change the NAME of an specific INPUT. The following names are available: USB, OPTIC, AES/EBU, SPDIF, STREAMER, PC, CD, DVD, SAT, DAT and MD.

After 10 seconds without interaction at the right knob, the display returns to normal operation and the parameter is stored. The countdown on the first row helps to control the time left.

Technical data

NAME

- The circuit of the **ZeroUno DAC-HPA** is based on the ESS SABRE³² ES9018S DAC chip plus a proprietary hybrid filter design and encapsulation against external noises.
- For optimum performance the SABRE chip runs on a proprietary implementation of special developed firmware
- One motherboard is based on a four-layer PCB with extra thick copper traces to achieve ultra-short signal paths with minimal wiring to avoid electromagnetic induction of noise and to insure perfect grounding
- Discrete power supplies built for the digital and the analogue sections using six toroidal transformers the last generation of ultra-low noise rectifier diodes and high quality low noise regulators
- Audio-grade resistors and capacitors
- Comprehensive noise regulation for all digital circuits
- Jitter free operation by patented high performance algorithm

Display example

- Integrated 32 bits volume control with a residual noise below -130dB for the best performances even at very low output levels
- On board output level setup to match different sensitivity levels of the direct connects power amplifies
- Voltage-compensated, ultra-low phase noise and low jitter crystal oscillators (clock) acts as master clock
- Two separated clocks for sampling families of 44.1, 88.2, 176.4, 352.8 kHz and 48, 96, 192, 384 kHz
- DAC output stage with the shortest possible signal path based on custom made amorphous audio transformers produced by Lundahl, with first order discrete analogue filter for the best THD and digital noise suppression
- True Class A discrete built analog output stage with zero negative global feedback
- Separately powered USB chip by a "quasi battery power supply" to avoid any distortion induced by the connected computer (no connection to the +5V powerline of the USB cable)
- USB input based on XMOS xCore audio chip with bit perfect transfer for 16bit, 24bit or 32bit data in PCM format up to 384 kHz, including support for native DSD/DoP and MQA.
- 4 Digital-Inputs: 1x USB 2.0; 1x true S/PDIF 75 Ohm BNC or true AES/EBU 110Ohm XLR; 1x S/PDIF RCA; 1x S/PDIF optical
 - USB input compatible with following audio formats via PC and MAC:
 - PCM: 44.1; 48; 88.2; 96; 176.4; 192; 352.8 and 384 kHz up to 32 bits
 - DSD (DoP): 2.822 MHz / DSD64, 3.072 MHz, 5.644 MHz / DSD128
 - MQA all formats
- SPDIF inputs (AES/EBU, BNC, RCA) compatible with PCM signals from 44.1, 48, 88.2, 96, 176.4 and 192 kHz, up to 24 bits. The optical input accepts music files with a resolution up to 24/96 kHz
- LCD Display with variable brightness and letter size for better reading from listening position
- Infrared remote control with direct function keys for volume, balance, absolute polarity, mute and input channel
- No drivers required for LINUX or MAC OSx
- USB Audio 2.0 driver available for Windows XP/Vista/7/8/10
- Gain = 26dB (x20);
- Zout < 1 ohm;
- Vo max = ±17V (24dBu)
- Pre-gain between 0dB to +24dB in step of 1dB
- Pout = 2.5Wrms onto 30ohm
- Headphone load within 30 to 600 ohm
- Power amplifier input impedance \geq 600ohm
- Distortion with 1KHz, 1Vrms, into 30ohm load: 2nd harmonic below -102db; 3rd harmonic below -100dB 4th harmonic below -125dB; 5th harmonic below -120dB

The Specifications in this document are subject to change without notice.

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