UC1 The Grimm Audio UC1 is a high quality Mastering and Mixing Universal Converter offering very broad digital application functionality. **This manual covers the specific functionality for use in an Audiophile System.**



UC1 - HiFi DAC application manual

Please read this manual before operating the unit.



Table of Contents

| 1. Introduction | 3 |
|----------------------------------|----|
| 2. Important Safety Instructions | 6 |
| 3. Installing | 8 |
| 4. Signal Connections | 10 |
| 5. Operation | 16 |
| 6. Menus | 26 |
| 7. Source Routing Examples | 26 |
| 8. Specifications | 33 |
| 9. Grimm Audio Limited Warranty | 34 |

1. Introduction

The Grimm Audio UC1 is a high quality Mastering and Mixing Universal Converter offering very broad digital application functionality. **This manual, however, covers the specific functionality for use in an Audiophile System.**

Please read this manual before using the UC1 to understand its operation and make optimal use of its versatility as a high audiophile quality DAC / HPA / PRE.

Applications Overview

Let's have a look at how to use the various inputs and outputs in this context.

DAC - The Primary Function

USB Connection :

The "rear" USB input [Input 6] as well as the "front" USB input, aka the "i-put" are connected to your digital source, be this a Server, a Streamer or a Computer.

AES3 S/PDIF Connection:

Two rear AES3 inputs can be connected to a digital source that has AES output, be this a Server or a Streamer.

The AES3 input can also connect a CD Transport via an AES > Coax adapter cable. Note that some high end CD Transports offer Master/Slave clock connection, such as the C.E.C. TL2N. The UC1's superior clock output can in this case can act as the Master when connected to the clock input of the Transport. The UC1's clock quality equals that of Grimm Audio's CC1 and CC2 master clocks.

When the UC1 is used as a standalone audiophile DAC and / or a monitor controller, the clock settings need to selected accordingly in the UC1's "F2" function menu.

PRE - Monitor Function

XLR Analog Loudspeaker Connection :

The UC1 offers 3 analog XLR outputs. One of these output pairs, preferably "Ana Out 3 | 4", can connect directly to a pair of active loudspeakers via their respective XLR inputs.

(Continued)

A secondary loudspeaker system can connected to a dedicated XLR output pair ["2nd Out"] that runs in parallel to the Headphone output situated on the front panel.

Digital RJ45 Loudspeaker Connection:

This dedicated connection connects directly to a pair of Grimm LS1 or PMC loudspeakers via their respective inputs.

In the menu you select to which Output the Loudspeaker system is connected. Although not typical in the Hifi Universe, please note that the UC1 offers several interesting Pro Audio monitor settings such as "Mono", "Side", and "Stereo Swap".

PRE - Gain & Control Functions

Gain / Volume / Mute / Polarity & Settings :

As the UC1 is a Pro Audio component, Gain - which can be increased or decreased via the volume dial knob - is expressed in positive or negative "dB deviation" from a reference loudness level nominally called "0dB". This level is by convention expressed in "db SPL". This level corresponds to an SPL of 79db in a small recording room. To adjust the UC1 to the HiFi usage convention, reset the default -8dB setting under the P2 function to -18db which effectively equals an output level of 0VU. The DAC's output clips at +18db (5Vpp), so the gain would cover a range of -18dB > +18dB.

Please note that the Volume and Mute functions are available for all input / output signal paths, but in the absence of a basic remote, these need to be adjusted manually. News : A remote control 3.5 mm socket on the back panel is for future use and is not yet enabled.

This is not really an issue as in almost all cases the UC1 as a standalone DAC will be connected to modern HiFi amplifiers with "remote". "Mute" is selected in "F1".

The signal's **Polarity** can be inverted in "P2", but note that this function is only available for Headphone Out signals traversing the "Foldback Route", which, if used, needs to be activated in "F1".

Furthermore, the 'Foldback' function is limited to two output paths, notably to Headphone and 2nd Out .

(Continued)

PRE - Headphone Function

6.35mm / 1/4" Jack Connection:

The UC1 offers a 6.3mm headphone output with gain control. The acoustic sensitivity of headphones on the market varies by as much as 30 dB. Consumer ear buds are more or less standardized on a sensitivity of approximately -14 dBu (~0.15V) for 100 dBA. Some professional 600 Ohm headphones however may need as much as 15 dBu (~4.4V) to reach 100 dBA. This example is of an HP with a sensitivity of 85dB/mw. <u>The UC1 is</u> <u>targeted at the professional audio market and therefore</u> <u>offers enough gain to drive these types of demanding</u> <u>headphones</u>. The majority of HiFi audio headphones are far less demanding and, as such, the UC1 will drive these with ease.

The Headphone output circuit runs through the Foldback route of UC1. As such, the Mute and Polarity Inversion functions are available. HP output volume can be controlled via the dial knob directly next to the output jack. Switching between HP and Loudspeaker monitoring can be done by briefly depressing the main dial at the far right of the front panel.

PRE - Analog In

XLR Connection:

This input was intended for recording analog signals for mixing / mastering purposes. It <u>does not</u> serve as a typical analog input to which first gain is applied before being"passed-through" to the UC1's analog outputs.

In effect, any Analog signal input would be converted by the A/D chip circuit to digital ... and then reconverted back to analog by the D/A chip circuit.

This may be useful only if the UC1 is connected to a pair of active loudspeakers, and where <u>it is not</u> connected to a Preamplifier > Amplifier > Loudspeaker system chain.

In extremis, It would be imaginable that if one wanted to connect a Turntable with a dedicated Line stage with integrated Phono to this analog input via RCA > XLR cable adapters, one could in fact do so. The SQ output post A/ D/A conversion would be "acceptable". This would also apply to the connection of a Reel-to-Reel tape machine.

2. Important Safety Instructions

Grimm Audio gaat er van uit dat u deze Engelstalige tekst volledig begrijpt. Als u hier moeite mee heeft dient u contact op te nemen met Grimm Audio. Op verzoek sturen wij u een vertaling toe.

Grimm Audio nimmt an, dass Sie diesen Englischen Text völlig verstehen. Wenn notwendig, nehmen Sie bitte Kontakt auf mit Grimm Audio. Auf Wünsch wird Ihnen eine Übersetzung zugeschickt.

Grimm Audio suppose que le lecteur comprend parfaitement le texte en Anglais ci-dessous. En cas de doute s.v.p. contacter Grimm Audio. Si necessaire, on pourra vous envoyer une traduction.

Grimm Audio da por supuesto que el texto en versión Inglesa no ofrece ninguna duda de interpretación y se entiende integramente. Si este no fuese su caso rogamos contacte con Grimm Audio quien, a petición, se encargaría de enviarle la correspondiente traducción.

Please follow these precautions when using this product:

- 1. Read these instructions.
- 2. Keep these instructions.
- 3. Heed all warnings.
- 4. Follow all instructions.
- 5. Dangerous voltage is inside this apparatus. Opening is only allowed by qualified service personnel.
- 6. Do not defeat the safety purpose of the polarized or grounding-type plug. A polarized plug has two blades with one wider than the other. A groundingtype plug has two blades and a third grounding prong. The wide blade or the third prong are provided for your safety. When the provided plug does not fit into your outlet, consult an electrician for replacement of the obsolete outlet.

- 7. Protect the power cord from being walked on or pinched, particularly at plugs, convenience receptacles, and the point where they exit from the apparatus.
- 8. Unplug this apparatus during lightning storms or when unused for long periods of time.
- 9. Do not use this apparatus near water.
- 10. Do not use this apparatus outside.
- Do not expose the apparatus to dripping or splashing. Do not place objects filled with liquids (flower vases, drink cans, coffee cups, etc) on the apparatus.
- 12. Clean only with a dry, soft, non-fluffy cloth. Do not spray any liquid cleaner onto the cabinet, as this may lead to dangerous shocks. Do not spray any liquid cleaner onto the faceplate, as this may damage the wooden front panel.

- 13. Install in accordance with the manufacturer's instructions.
- Do not install near any heat sources such as radiators, heat registers, stoves, or other apparatus (including amplifiers) that produce heat. Avoid exposure to direct sunlight.
- 15. Use only attachments or accessories specified by the manufacturer.
- 16. Use only with a cart, stand, bracket, or table designed for use with professional audio or music equipment. In any installation, make sure that injury or damage will not result from cables pulling on the apparatus and its mounting. If a cart is used, use precaution when moving the cart/apparatus combination to avoid injury from tip-over.
- 17. This apparatus typically runs warm when operated. Install in a normal ventilated area. If the product will be used in a rack, make certain there is sufficient air movement within the rack. Preferably offer some

empty rack space above the unit and do not place it on top of hot equipment.

- 18. Refer all servicing to qualified service personnel. Servicing is required when the apparatus has been damaged in any way, such as when the powersupply cord or plug is damaged, liquid has been spilled or objects have fallen into the apparatus, the apparatus has been exposed to rain or moisture, does not operate normally, or has been dropped.
- 19. WARNING: To reduce the risk of fire or electric shock, do not expose this apparatus to rain or moisture.

3. Installing

Unpacking and Inspection

Your UC1 was carefully packed at the factory and the packaging it came in was designed to protect it from the trials and tribulations of shipping. Keep the box and all packing materials, so that in the unlikely event that you need to return the UC1 for servicing, you can do so safely.

AC Power Hookup

The UC1 has a 'full mains' switch mode power supply that works with all choices of mains voltage world wide. This SMPS is developed by Grimm Audio and offers the highest possible sound quality. For your safety and for optimal sound quality, always use a power outlet with proper safety earth.

The UC1 is shipped with a mains cable. Grimm Audio cannot be held responsible for problems caused by using the UC1 with improper AC wiring.

The UC1 does not have a power switch on the front panel, it needs to be powered down with the mains switch at the back. A convenient way to power down the unit is to use a power strip equipped with a switch, or a power sequencer.

The high sound quality DC coupling of the UC1's analog outputs may result in a popping sound when turning off the unit while the downstream monitor system is still turned on. We recommend to use a manual or automatic power up and power down sequence as follows: first turn the UC1 on and then the monitors. And at the end of your working day, first turn off the monitors and then the UC1.

Mounting the UC1

Since high quality audio electronics draw above average current, the UC1 runs warm. Note that the bottom plate is used as a heat sink, there is no active cooling. Please allow for sufficient air movement around the unit. The UC1 does not produce strong RF fields nor is susceptible to them. You can position it near other digital gear such as computers and servers / streamers without worry.

Grimm Audio products have a real wooden face plate that provides a beautiful and vivid appearance. The UC1 front panel is made of Padauk. The panel is finished with an environmentally friendly oil. To maintain the outstanding looks, one is advised to take some precautions:

- Use a dry woolen cloth to clean the wood.
- Do not place the UC1 in very dry environments.
- Do not use chemical or alcohol based cleaner on the wood.
- Avoid exposure to direct sun light, since the Padauk wood is susceptible to discoloration in UV light.

4 A. Signal Connections - Front Panel



4 B. Signal Connections - Back Panel



On the back panel

The USB connector on the back offers a USB Audio Class 2 compliant interface.

This manual discusses operation of the UC1 as a high quality multi-input DAC. In this application the clock source needs to be selected in the UC1 menu instead of the computer OS. Please refer to the Menu chapter, paragraph 'F2', for instructions.

All analog inputs and outputs on the back of the UC1 are XLR's. Our design aim was to avoid sub-D connectors since they are not of the same quality as XLR's. The nominal output voltage of all analog connections is +4 dBu. This is aligned to -14 dBFS, hence the clip level is at +18 dBu. These are common levels in mastering studios. The analog inputs on the back can optionally be switched to a clip level of +24 dBu so one can drive vintage outboard gear to the max without clipping the UC1 input that receives their signal. Of course the 1:1 relation between output and input level of the UC1 is then lost.

The digital inputs and outputs are in AES3 format. Leftmost is the "LS1 control out' cat5 connector. This is not ethernet but a proprietary combination of AES3 and midi control data. We share this type of connector with our friends of PMC. If you own a digital PMC loudspeaker such as the TwoTwo, IB2S or QB1 series, the UC1 can be setup to send the right control data and control its volume and mute function. To do this, please follow these instructions:

Boot the UC1 as PMC loudspeaker controller

Directly after turning on the UC1 a progress bar startup display can be seen on the UC1 screen. While this bar is visible, first press the top button, hold it and then press the bottom button. Release when the bar is gone. Your UC1 is now in 'PMC mode'. In case you like to return to 'Grimm LS1' mode, reset the unit to factory default by only pressing the top button during startup.

(Continued)

The word clock in- and outputs are 75 Ohm terminated. The UC1 clock circuitry is of similar reference quality as that of our CC1 and CC2 master clocks.

The UC1 can slave to word clock, Digital in 1 and Digital in 2. This enables you to monitor digital sources that cannot be slaved to the UC1 word clock output such as CD players.

Note that S/PDIF sources can be connected via an S/PDIF to AES adapter cable (that can be sourced from Grimm Audio).

The PLL that synchronizes the UC1's internal clock to the external clock is of a very special kind. It achieves lock very fast and then gradually increases its jitter suppression down to very low jitter frequencies, effectively only following very slow drift of the source clock. This proprietary 'Hybrid PLL' technology offers the best of both worlds: instant lock so there is no gap when switching sample rates, and high jitter suppression as well. Standard PLL's will either lock fast or offer high Jitter suppression but not to both at the same time. If you like to record from both inputs simultaneously, please make sure that all devices that are not used as clock source are synchronized by means of the UC1 word clock or AES outputs.

The remote control 3.5 mm jack socket on the far right is for future use and not yet enabled.

On the front panel

A unique feature of the UC1 is its front USB input, the 'iput'. It is intended to connect a secondary stereo USB source such as a laptop or smart device to the UC1.

Smart devices, such as an iPAD, are often connected wireless to a computer. The advantage in using the i-put with a smart device is that, in contrast with wireless Connections, all sample rates and bit depths are supported and data is transferred bit transparent. It may also offer you a more reliable connection.

To use the USB front i-put, please note these two important instructions :

1. The UC1 sample rate cannot be controlled from the front USB. As a result the UC1 can only play audio from the front USB input if the playback sample rate of the device is the same as the current UC1 sample rate. In case the device playback sample rate does not match the UC1 sample rate, the i-put meters in the main and secondary displays are red lined and the sound is muted.

2. If you like to connect an Apple smart device such as an iPhone or iPad, a special conversion cable called "Lightning to USB3 camera adapter cable" or "USB-C to USB adapter cable" (depending on your device) is needed. These cables carry a chip that enables audio playback to USB from iOS devices. Some cables also provide a means to power the iDevice externally.

For the rest, the front USB behaves like that on the back.

The front analog input is a stereo unbalanced 6.3mm jack with a gain control next to it. It can be used for

(Continued)

instance when someone brings in an analog MP3 player with a reference track. This input has a high input impedance of 1 Mohm.

Leftmost on the front is a 6.3mm headphone output with gain control. The acoustic sensitivity of headphones on the market varies by as much as 30 dB. Consumer ear buds are more or less standardized on a sensitivity of approximately -14 dBu for 100 dBA. Some professional 600 Ohm headphones however may need as much as 15 dBu to reach 100 dBA. The UC1 is targeted at the professional audio market and therefore offers enough gain to drive these headphones.

Please be aware that with sensitive headphones the UC1 can produce sound pressure levels that can cause immediate and serious hearing damage! Grimm Audio does not take any liability to hearing loss when using the UC1. Please act responsibly and protect yourself!

5. Operation

The UC1 functions are controlled with the main dial on the right in combination with the two small push buttons next to it. But before we look at that in more detail, let's first walk through the internal routing of the device.

General Routing Overview

Figure 2 on the next page shows <u>the entire routing</u> <u>diagram</u> of the UC1.

All 5 inputs [10channels] are routed to their respective USB channels transparently without any form of processing. Similarly all 8 USB channels from the main USB 1 input flow unaltered to the outputs on the UC1 except for the 1 USB input [2 channels],aka the "i-put" that is associated with the main output.

The yellow blocks in figure 2 show the monitor path of the UC1. All I/O channels can be routed to this monitor path. They can be auditioned in their original stereo form, or in mono (L+R), side (L-R) or in swapped channel order form.

They then receive digital Volume Control & Mute - applied to both channels, or to L or R Mute.

'Hidden functions' of the UC1

- Press the main dial briefly to mute the main speakers and turn on the headphone output. Press briefly again to return to the main speakers.
- Press-hold the main dial until the indicated source starts blinking. Then turn to change the monitored source. Release to select. Alternatively, release the dial when the blinking starts. Then turn to change source. Press briefly to select.
- Press-hold the top control button to enter the metering display. Press-hold again to exit.
- Press-hold the bottom control button to exit or enter Foldback mode, when enabled via the menu.
- Check chapter 4 Signal Connections, "On the back" for instructions about how to set the cat5 output to PMC digital monitor control or reset the UC1 to factory default settings.



Buttons and display

Most functions of the UC1 are adjusted via the main dial and the two control buttons on its left side. Additionally there are knobs for the phones output gain and the jack input gain.

The main dial can be turned and pushed. In normal operation, turning changes the gain of all monitor paths (main, 2nd and phones) and pushing mutes the speakers while engaging the headphones output on the front. When press-holding the main dial, the current monitor source (indicated top left of the display) starts



Main screen, USB 1/2 selected

blinking. Keep pressing the main dial while turning it to change the monitor source. All inputs and outputs can be selected. Their name is shown in the display and their channels are highlighted in the meter. When releasing the dial the indicated source stays selected. Alternatively, release the dial when the blinking starts. Then turn to select a source and press to confirm.

The large number in the display indicates the current monitor gain level in dB's. The monitor outputs offer 8 dB of clean gain above '0' on the display. In other words: a full scale signal will only reach full scale on a monitor output if the monitor gain is set to +8 dB. This may seem odd, but the idea is that the '0' indication of the display is your reference playback level to which you always return after a louder of softer playback. This is common practice in mastering. The '0' level is usually aligned with an SPL of 79dBC when playing a SMPTE reference noise signal on a single channel. The Grimm LS1 is factory aligned to that, you may adjust the gain of your 3rd party amplifier to achieve the same result - or to select another acoustic SPL that suits you as reference playback level. In case you feel the need to adjust the '0' level of the UC1, you can change it in menu P2 (see chapter 6. Menus).

Note that the monitor gain can be set to display values up to +18, which means that there is 10 dB of positive gain available in the UC1 to audition very soft noises. Note that any signal above -10 dBFS will in that case clip – please use this extra gain with caution!

The volume control is in 0.5 dB steps above -17 dB and 1 dB below it. If you have a Grimm Audio LS1 or PMC digitally controlled loudspeaker connected to the proprietary cat 5 output on the back, the volume control



Main screen, Digital 1/2 selected, in phones mode

takes place in the DSP of the loudspeaker. In all other cases the volume control is in the DSP of the UC1. At volume setting '+8' the data to the digital output and build-in DACs is put through transparently. At other levels the digital gain control is properly dithered at 24 bits.

Optionally, volume control of the main out can be disabled in the menu. In that case the indicated monitor gain only applies to the 2nd out and phones.

The knob to the right of the phones jack can be used to align the headphones' acoustic level with that of your main monitor system. The idea is to switch between speakers and phones (removing your headphones when listening to the speakers) and adjust this control until speakers and phones sound equally loud to you. From then on just use the main dial for changing your monitor gain and leave the phones control at its calibrated level.

In the bottom left corner of the display the clock status of the UC1 is indicated. The clock source can be set manually. Please refer to the menu chapter for instructions.

Updating your UC1 firmware

Occasionally we will release an update to the firmware of your UC1. To keep informed about firmware updates, please subscribe to our newsletter at the footer of our website pages. In the downloads section of the UC1 web page at grimmaudio.com the latest firmware can be downloaded.

First check the current firmware version of your UC1 to make sure an update is needed.

For Windows, or Macintosh please refer to the UC1's main manual for the procedures to follow for updating the firmware.

Reset your UC1 to Factory Defaults

In case your UC1 shows odd behavior or when you like to restore all menu items to their original settings, you may restore the factory defaults by pressing the top button when the progress bar is visible during power up.

6. Menus

The UC1 has many functions, preferences and system setup items that can be controlled from the menu. To enter the menu, press the top button with 'menu' next to it on the display. You will see a screen with 6 tabs, labeled F1 and F2 for Functions, P1, P2 and P3 for Preferences and S1 for System info. We walk you through all these settings in the pages below. To move to the next tab, press the top button 'next tab'. To exit the menu, press the bottom button 'exit'. Note that the exit button is disabled when an option is selected and its arrow is blinking. Turn the main dial to select a function, press to change it, then turn to browse the options and press again to confirm your choice.

F1 menu

Whenever you enter the menu, the F1 page will be available first. This page gives access to monitor functions that are often used. • Monitor source. You can select all inputs and outputs here by pressing the main dial briefly (the arrow starts blinking) and then turning the dial. When you reach the last source, the order wraps and the first source appears again. To confirm a choice, press the dial again (the arrow stops blinking). This item does the same as press-hold the main button and then turn. The result is immediately visible in the left top corner of the display, and the selected source becomes audible.

Grimm LS1 customers, please note that the UC1 cannot select the analog and digital inputs of the LS1 as a source. In case you need to access these, please use the

(Continued)



F1 menu.

LS1 PC Remote software with the UC1. It is not possible to combine the UC1 with the LS1r hardware controller and LS1i USB interface. The UC1 replaces these.

• Special. After a press of the dial the choices are stereo / mono / side / swap. Note that mono = L+R and side = L-R (they are also known as 'M' and 'S'). "Swap" flips the left and right channels . At the end of the dial the order wraps back to stereo again. This makes it possible to AB Compare stereo & mono, mono & side and swap & stereo. If you confirm a choice by pressing the dial once more, the arrow stops blinking. If you then exit to the main screen the function is shown in the screen (except for 'stereo', which is the normal state). • Foldback. When in the main screen - as a quick access feature - a single press of the bottom button turns off and on foldback (in case a foldback source was selected in the menu).

• Mute. To mute the loudspeakers from the main screen you just press the main dial. This turns off the selected speakers and turns on the phones at the same time: it toggles between speakers and phones.

F2 menu

• Clock. This setting is only available when the back USB is not connected to a computer (normally the USB host controls the UC1 clock). It allows you to choose all sample rates from 44.1 to 192 kHz in master mode, or to slave to word clock, digital in 1 or digital in 2.



F2 menu.

• Autoswitch clock. This setting influences the UC1 behavior when in slave mode. When 'yes' the UC1 clock rate will follow any change of the selected clock source. When 'no', the UC1 will maintain the clock rate that it found when first slaving to the selected clock source. If the source then switches sample rate, the clock indication on the display will turn red and start flashing to indicate it lost lock. The 'no' mode is helpful when you are running a project with a specified sample rate and it should keep running at that rate, even when by accident the source that you sync to changes sample rate.

• 2nd out. This flips the monitor path to the 2nd speaker output XLR's on the back (main and phones are muted).

• Compare mode. This mode is used when comparing two sources at equal loudness, for instance when evaluating the impact of a compressor/limiter in the analog loop. Usually the external processor will have changed the signal's loudness and when flipping between the two sources, this loudness jump hampers a proper quality judgement. In compare mode the volume setting is memorized for every source individually instead of global. So if you listen to the first source, adjust the volume to taste, then switch to the second source and adjust that volume to a similar level, flipping between the two sources can be done at equal loudness. To indicate that compare mode is engaged, the volume number turns green.

P1 menu

The P1 menu is the first of three Preference menus. These are settings that are usually only changed to tailor the UC1 to your system setup.



P1 menu.

• Main out. Here you select your main monitor output. There are three options: LS1 / PMC on cat5, Dig out 1|2 and Ana out 1|2. The selected output will receive volume control and monitors the audio from the monitored sources. Their playback will be interrupted when another monitor source is chosen.

• Max in Analog 1 | 2. This preference let's you choose between 18 dBu or 24 dBu as maximum modulation (0 dBFS) input voltage. The default is 18 dBu (all UC1 outputs have a maximum modulation at 18 dBu). In case a vintage device is used into your analog input, and that you like to drive to maximum swing, you can change this setting to 24 dBu.

• Main out volume. By default it is turned on. Some people prefer to use an external analog volume control in stead of the internal digital one. They can disable the main out volume control here. Volume control of 2nd out and headphones will still work.

• PC control (midi/volume). Not Applicable to this application.

P2 menu

• Monitor mode. In 'normal' mode the three monitor outputs (main, 2nd speaker and phones) alternate. Only one is active at a time. In 'split' mode the 2nd speaker and phones paths are decoupled from the monitor system and become an independent path for USB 9&10, without any digital volume control. The 2nd out and phones channels carry the same signal and are always on. The rightmost channels in the meter display now show the USB 9&10 levels in stead of the monitor output levels.

• Soft key. Here the function of the bottom button in the main screen can be selected. You can choose between: *2nd out* (switch to 2nd speaker output); *special* (enable the selected 'special' function from menu F1, like mono or side); *dim 20 dB* (attenuate the output by 20 dB); *compare mode* (enable the compare mode of menu F2); *fast 0 dB* (set the volume level to the current reference level).

• Phones phase. This flips the polarity of the headphones output. Since there is no official standard for the relationship between the electrical signal polarity and the acoustical pressure polarity of headphones, a positive going electric pulse can result in a pressure increase on some brands' headphones and a pressure decrease on others. The UC1 polarity switch can be used to achieve a positive polarity with your microphone / headphones combination so that an increase in pressure in front of your microphone will result in an increase in pressure inside your headphones.

(Continued)



P2 menu.

Since headphones always show some leakage from environmental sounds, selecting the right polarity will improve the coherence of the leaked acoustic sound.

• Ref level. When this line is selected, a press on the main dial stores the current volume level as the '0 dB' reference playback level. The default setting is -8 dB, which means that the UC1 monitor output has 8 dB of gain available above the reference setting. If you need more or less 'excess gain', you may change it here.

P3 menu

- Display brightness. This function controls the brightness of the display from 1 to 10.
- Display auto dim. If auto dim is turned on the display will turn off a few seconds after a control is used.

• Main screen meters. This enables or disables the presence of a simple 3-stage meter on the main screen.

S1 menu

This menu shows the hardware and firmware versions of your UC1. It also shows if the UC1 is booted in LS1 mode or PMC mode.



P3 menu.



S1 menu.

7. Source Routing Examples

Suggested Routing & Settings - Source Specific

This section covers suggested routing and settings for 4 Digital Source options - 3 server / streamers, and a CD Transport. A brief description of each is shown below, and is illustrated by a diagram on pages 29-32.

| A. Routing USB1 Server/Streamer - Rear Panel | C. Routing Grimm MU1 Server/Streamer |
|---|--|
| Figure 2a on page 29 shows the routing diagram of the UC1 for the USB1 input connection on the rear panel. | Figure 2c on page 31 shows the routing diagram of the UC1 for the Grimm MU1 connected to AES3 Dig 1/2. |
| The selected output shown is Ana 3 4 which bypasses all functions and controls. To allow all functions and controls, select output Ana 1 2 (aka Main) or "2 nd Out". | The selected output shown is "2 nd Out", and all controls and functions are enabled. If "Phones" and "polarity" are not desired, select output Ana 1 2. |
| | |
| B. Routing USB I-put Server/Streamer - Front Panel | D. Routing CD Transport |
| Figure 2b on page 30 shows the routing diagram of the UC1 for the USB "I-put" connection on the front panel. | Figure 2d on page 32 shows the routing diagram of the UC1 for a CD Transport connected to AES3 Dig 2/2, |
| The selected output shown is "2 nd Out", and all controls and functions are enabled. If "Phones" and "polarity" are not desired, select output Ana 1 2. | The selected output shown is "2 nd Out", and all controls and functions are enabled. If "Phones" and "polarity" are not desired, select output Ana 1 2. |

A. Routing USB 1 | Back



| USB | 81 | F1 se | ettin | gs | | |
|------|----|-------|-------|------|-------|--|
| SB 1 | | -12.0 | | next | : tab | |
| | | | | | | |

| F1 | F2 | P1 | P2 | P 3 | S1 | |
|--------------------|--------|-----------|-----------|------------|-----------|--|
| > monitor source : | | | | USB 1 | | |
| special : | | | ster | reo m | ono | |
| foldback : | | | 0 | on of | f | |
| | mute : | | | on of | f | |
| ma | ster 4 | 4.1 | exit | | | |

USB1 | F2 settings

| USB 1 | | -12.0 | | next tab | |
|-------------------|-----------|-----------|----------|------------|-----------|
| F1 | F2 | P1 | P2 | P 3 | S1 |
| clock master 44.1 | | | | | |
| autos | witch o | clock : | yes no | | |
| 2 | nd out | : | on off | | |
| compare mode : | | | on off | | |
| master 44.1 | | | | exit | |

Settings

- 1. Using the main VOL / Source Dial, select the USB1 input under F1.
- 2. Via the two setting buttons, scroll through and select the F1, F2, functions and the P1 and P2 Preferences as shown.
- 3. The P2 "reference level" can preferably be between -8 and -18 ... e.g. -12
- 4. The Clock master setting for this source's playback will automatically set itself to the source clock when "autoswitch clock" under F2 is set to "yes".

USB1 | P1 settings

| USB 1 | | -12.0 | | next tab | |
|----------------------|---------|-----------|-----------|-------------------|-----------|
| F1 | F2 | P1 | P2 | P 3 | S1 |
| > main out : Ana Out | | |)ut 3 4 | 2 nd O | ut 1 2 |
| 1 | max In | Analo | g 1 2 : | 18 dBu | l |
| main | out vol | ume : | on off | | |
| compare mode : | | | on off | | |
| master 44.1 | | | exit | | |

USB1 | P2 settings

| USB 1 | | -12.0 | | next tab | |
|------------------|----|---------------------|-------------------|-----------|-----------|
| F1 | F2 | P1 | P2 | P3 | S1 |
| > monitor mode : | | normal | | | |
| softkey : | | 2 nd Out | | | |
| phones phase : | | | normal inverted | | |
| ref level : | | | 0dB - 12.0dB | | |
| master 44.1 | | | | exit | |



| i- | put l | USB | F1 set | ttings | |
|--------------------|-------|---------------|-----------|------------|-----------|
| i-put l | JSB | -12 | 2.0 | next | t tab |
| F1 | F2 | P1 | P2 | P 3 | S1 |
| > monitor source : | | i-put USB | | | |
| special : | | stereo mono | | | |
| fold | lback | | (| on of | f |
| mute : | | | on off | | |
| master 44.1 | | | | exit | |

| i-put USB F2 settings | | | | | |
|-------------------------|-----------|-----------|-----------|------------------------|-----------|
| i-put | USB | -12 | -12.0 | | t tab |
| F1 | F2 | P1 | P2 | P3 | S1 |
| clock master 44.1 | | | | | |
| autos | witch o | clock : | | y <mark>es</mark> no |) |
| 2 | nd out | : | C | on of | f |
| compare mode : | | | on off | | |
| master 44.1 | | | | exit | |

Settings

- 1. Using the main VOL / Source Dial, select the "I-Put USB" input under F1.
- 2. Via the two setting buttons, scroll through and select the F1, F2, functions and the P1 and P2 Preferences as shown.
- 3. The P2 "reference level" can preferably be between -8 and -18 ... e.g. -12
- 4. The Clock master setting for this source's playback will automatically set itself to the source clock when "autoswitch clock" under F2 is set to "yes".

i-put USB | P1 settings

| | - | | | | |
|----------------------------|---------|-----------|---------|-------------------|-----------|
| i-put | USB | -12.0 | | nex | t tab |
| F1 | F2 | P1 | P2 | P3 | S1 |
| > main out : Ana (| | |)ut 1 2 | 2 nd O | ut 1 2 |
| max In Analog 1 2 : 18 dBu | | | | | |
| main | out vol | ume : | C | n of | f |
| com | pare m | ode : | | on of | |
| ma | nster 4 | 4.1 | | exit | |
| i-put USB P2 settings | | | | | |
| i-put USB -12.0 | | | 2.0 | nex | t tab |

| I-put USB | | -12 | .0 | nex | lan | | |
|------------------|-----------|-----------|---------------------|------|-----|--|--|
| F1 | F2 | P1 | P2 P3 S1 | | | | |
| > monitor mode : | | | normal | | | | |
| softkey : | | | 2 nd Out | | | | |
| pho | nes ph | ase : | normal inverted | | | | |
| ref level : | | | 0dB - 12.0dB | | | | |
| ma | ster 4 | 4.1 | | exit | | | |



| Dig In 1 2 F1 settings | | | | |
|--------------------------|------------|---------------|------------|-----------|
| Dig In 1 2 | -12 | -12.0 | | t tab |
| F1 F2 | P1 | P2 | P 3 | S1 |
| > monitor source : | | Dig In 1 2 | | |
| special : | | stereo mono | | |
| foldback | : : | (| on of | f |
| mute : | on off | | | |
| master 4 | | exit | | |

Dig In 1|2 | F2 settings

| Dig In 1 2 | | -12.0 | | next tab | | |
|--------------------|-------------------|-----------|----------|------------|-----------|--|
| F1 | F2 | P1 | P2 | P 3 | S1 | |
| | clock master 44.1 | | | | | |
| autoswitch clock : | | | yes no | | | |
| 2nd out : | | | on off | | | |
| compare mode : | | | on off | | | |
| master 44.1 | | | | exit | | |

Settings

- 1. Using the main VOL / Source Dial, select the Dig 1 | 2 input under F1.
- 2. Via the two setting buttons, scroll through and select the F1, F2, functions and the P1 and P2 Preferences as shown.
- 3. The P2 "reference level" can preferably be between -8 and -18 ... e.g. -12
- 4. The Clock master setting for this source's playback will automatically set itself to the source clock when "autoswitch clock" under F2 is set to "yes".

Dig In 1|2 | P1 settings

| | • | | | | |
|--------------------|--------------------|-----------|----------|---------------------------------|-----------|
| Dig In 1 2 | | -12.0 | | next tab | |
| F1 | F2 | P1 | P2 | P3 | S1 |
| > mai | > main out : Ana (| | | Out 1 2 2 nd Out 1 2 | |
| max In Analo | | | g 1 2 : | 18 dBı | ı |
| main out volume : | | | on off | | |
| compare mode : | | | on off | | |
| master 44.1 | | | | exit | |

Dig In 1|2 | P2 settings

| Dig In 1 2 | | -12.0 | | next tab | |
|------------------|----|-----------------------|-----------|-----------|-----------|
| F1 | F2 | P1 | P2 | P3 | S1 |
| > monitor mode : | | normal | | | |
| softkey : | | 2 nd Out | | | |
| phones phase : | | normal inverted | | | |
| ref level : | | 0dB - 12.0dB | | | |
| master 44.1 | | | exit | | |



| Dig In 2 2 F1 settings | | | | | | | |
|--------------------------|----|-------------|----------|------------|-----------|--|--|
| Dig In 2 2 -12 | | 2.0 | next tab | | | | |
| F1 | F2 | P1 | P2 | P 3 | S1 | | |
| > monitor source : | | Dig In 2 2 | | | | | |
| special : | | stereo mono | | | | | |
| foldback : | | (| on of | f | | | |
| mute : | | on off | | f | | | |
| master 44.1 | | | exit | | | | |

| Dig In 2 | 2 | F2 settings |
|----------|---|-------------|
|----------|---|-------------|

| Dig In 2 2 | | -12.0 | | next tab | |
|--------------------|-----------|-----------|-----------|------------|-----------|
| F1 | F2 | P1 | P2 | P 3 | S1 |
| clock master 44.1 | | | | | |
| autoswitch clock : | | | yes no | | |
| 2nd out : | | | on off | | |
| compare mode : | | | on off | | |
| master 44.1 | | | | exit | |

Settings

- 1. Using the main VOL / Source Dial, select the Dig 2 2 input under F1.
- 2. Via the two setting buttons, scroll through and select the F1, F2, functions and the P1 and P2 Preferences as shown.
- 3. The P2 "reference level" can preferably be between -8 and -18 ... e.g. -12
- 4. The Clock master setting for this source's playback will automatically set itself to the source clock when "autoswitch clock" under F2 is set to "yes".

Dig In 2 | P1 settings

| Dig In 2 2 | | -12.0 | | next tab | |
|--------------------|----|-----------|----------|-------------------|-----------|
| F1 | F2 | P1 | P2 | P3 | S1 |
| > main out : Ana (| | |)ut 1 2 | 2 nd O | ut 1 2 |
| max In Analo | | | g 1 2 : | 18 dBı | ı |
| main out volume : | | | on off | | |
| compare mode : | | | on off | | |
| master 44.1 | | exit | | | |

Dig In 2|2 | P2 settings

| Dig In 2 2 | | -12.0 | | next tab | |
|--------------------|----|---------------------|-------------------|-----------|-----------|
| F1 | F2 | P1 | P2 | P3 | S1 |
| > monitor mode : | | normal | | | |
| softkey : | | 2 nd Out | | | |
| phones phase : | | | normal inverted | | |
| ref level : | | | 0dB - 12.0dB | | |
| master 44.1 | | | exit | | |

7. Specifications

- Analogue output specifications
 - Clip level: +18 dBu
 - THD+N: -100 dB
 - SNR: 114 dB
 - Output impedance: <25 Ohm below 20 kHz
 - Headphone 6.3 mm stereo jack max 15 dBu in 600 Ohm (30 mW)
- Analogue input specifications
 - Clip level: XLR +18 or +24 dBu, Jack -18 dBu to +18 dBu
 - THD+N: -105 dB
 - SNR: 115 dB
 - Input impedance: 20 kOhm below 20 kHz
 - 6.3mm stereo jack input impedance 1MOhm
- Digital input and output specifications
 - AES3 on XLR
 - Impedance 110 Ohm
 - Sensitivity 200mVpp 5Vpp

- Clock specifications
 - Sample rates: PCM 44.1 kHz 192 kHz
 - Internal intrinsic clock jitter : <0.6 ps RMS (> 10 Hz)
 - Clock frequency master mode: 44.1 or 48 kHz ± 20 PPM, -5 +50 °C
 - BNC Word Clock 75 Ohm input and output impedance, 5V
- Chipset
 - D/A : AKM AK4396VF | A/D : TI PCM4202
- General specifications
 - Power supply voltage: 90-240VAC, 50/60Hz
 - Power consumption: 30 W
 - Fuse: F630 mA
 - Maximum ambient temperature for operation: 40 °C
 - Life expectancy power supply electrolytics > 45.000hr
 - Weight: 3.5 kg
 - Dimensions: 435 x 230 x 44 mm (ex. rack ears)
 - Wood type of front panel: Padauk

8. Grimm Audio Limited Warranty

Grimm Audio BV ("Grimm Audio") warrants this product to be free of defects in material and workmanship for a period of two (2) years for parts and for a period of two (2) years for labor from the date of original purchase. This warranty is linked to the serial number of the device and can be transferred to second hand owners if they can show their purchase bill.

The original owner can extend his limited warranty to a period of five (5) years for labor and parts if he becomes member of the Grimm Audio Community and sends his purchase details to the Grimm Audio factory via the webform at grimmaudio.com/warranty. The extended warranty is enforceable only by the original retail purchaser and cannot be transferred or assigned.

During the warranty period Grimm Audio shall, at its sole and absolute option, either repair or replace free of charge any product that proves to be defective on inspection by Grimm Audio or its authorized service representative. In all cases disputes concerning this warranty shall be resolved as prescribed by law. To obtain warranty service, the purchaser must first call or write Grimm Audio at the address and telephone number printed below to obtain instructions where to send the unit for service. All enquiries must be accompanied by a description of the problem. All authorized returns must be sent to Grimm Audio or an authorized Grimm Audio repair facility postage prepaid, insured and properly packaged. Proof of purchase must be presented in the form of a bill of sale or some other positive proof that the product is within the warranty period. Grimm Audio reserves the right to update any unit returned for repair. Grimm Audio reserves the right to change or improve design of the product at any time without prior notice.

This warranty does not cover claims for damage due to abuse, neglect, alteration or attempted repair by unauthorized personnel, and is limited to failures arising during normal use that are due to defects in material or workmanship in the product. In no event will Grimm Audio be liable for incidental, consequential, indirect or other damages resulting from the breach of any express or implied warranty, including, among other things, damage to property, damage based on inconvenience or on loss of use of the product, and, to the extent permitted by law, damages for personal injury. © 2021, Grimm Audio BV. All rights reserved

Grimm Audio UC1 manual v1.1.3v2, 6/12/21

Reproduction in whole or in part is prohibited. Specifications subject to change without notice.

Grimm Audio, Eindhoven, The Netherlands General questions: info@grimmaudio.com Support questions: support@grimmaudio.com Tel. +31 40 213 1562





Grimm Audio BV The Netherlands +31 40 213 1562 info@grimmaudio.com

www.grimmaudio.com