

dCS Scarlatti DAC
Stereo Digital to Analogue Converter

User Manual
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USING THE *dCS Scarlatti* DAC FOR THE FIRST TIME

Congratulations on purchasing your *dCS Scarlatti* DAC.

Before using your unit, please read this section and the Step by Step Guide. This will enable you to set the unit up quickly and safely with your hi-fi system.

From time to time, *dCS* will release updated software on CD that you can install yourself using the **CD Update** feature. Please check our web-site occasionally to see if new *Scarlatti* software is available, or consult your dealer.



Software version 1.30 adds *dCS*-encrypted DSD, 24/352.8 and 24/384 in Dual AES format. This allows the *Scarlatti* DAC to be used in DSD mode or DXD mode with a *Vivaldi Transport* (which has no 1394 interface) – see page **12**.

What's in the box?

Check that the box contains the following items:

- *dCS Scarlatti* DAC
- Manual and Menu / Setup Guide
- Power cable
- 1394 cable
- BNC cable
- 2 spare fuses

Notify your dealer as soon as possible if anything is missing or damaged. We suggest that you retain the original packaging for possible future use. If this is not possible, replacement packaging can be ordered from *dCS* or our distributors. Details can be found on our web site at www.dcsltd.co.uk.

The *Scarlatti Transport* is supplied with a programmed Nevo Q50 remote control. If you do not have a *Scarlatti Transport* in your system, a programmed Nevo Q50 can be ordered direct from *dCS* as an optional extra.

Positioning the unit

Units in the *Scarlatti* range are designed to be mounted on separate shelves of a rack and must NOT be stacked directly on top of each other. Place each unit on a firm, vibration free base, allowing convenient connection to the other parts of your system. To prevent overheating, we recommend that you leave some free space around the unit to allow for ventilation.

Safety Notice



Your *dCS Scarlatti DAC* contains no user serviceable parts. **DO NOT** attempt to open the case as there are potentially dangerous voltages present inside. In the event of the unit developing a fault, please contact your dealer in the first instance.

To maintain protection from electric shock, the unit **MUST** be connected to mains earth (ground) via the power cable. Also, unearthed systems do not give the best sonic performance.



This product is lead-free and complies with the RoHS directive.

Before connecting the power cable to the unit for the first time, please check that it has been set to the correct operating voltage for your mains supply. The unit's voltage setting is shown on the serial number label. If this does not match your local supply voltage, **DO NOT** attempt to use the unit. Contact your dealer to have the unit reset. Using the unit with the wrong mains setting for your local supply may result in serious damage to the unit and will invalidate the warranty. Do not attempt to reset the voltage yourself.

We do not recommend the use of mains regenerators. However, if you do wish to use a mains regenerator with variable voltage and frequency, we recommend that you set the voltage to match your local voltage and the frequency to either 50Hz or 60Hz **ONLY**.



Damage caused to your *Scarlatti DAC* by misuse of a mains regenerator or by a malfunctioning mains regenerator is not covered by the warranty.



Disposal at end-of-life: the symbol indicates that this product should not be treated as normal household waste. It should be recycled, so please take it to an approved collection facility.

STEP-BY-STEP GUIDE

This section guides you through setting up the unit for basic operation.

Preliminaries

The Menu / Setup Guide sheet details the menu structure and details the two most common set-ups.

For digital interfaces, use with cables designed for digital audio:

- for AES/EBU interfaces use 110Ω screened, twisted pair cables fitted with one male XLR connector and one female XLR connector.
- for SDIF, Word Clock or SPDIF BNC interfaces, use 75Ω coax cables fitted with BNC plugs.
- for SPDIF RCA interfaces, use 75Ω coax cables fitted with RCA Phono plugs.
- for TOSLINK optical interfaces, use Toslink fibre-optic cables.
- for IEEE 1394 interfaces, use the IEEE 1394 cable provided with the unit.

For analogue outputs, use with screened cables of the correct type:

- for balanced outputs, use screened, twisted pair cables fitted with one male XLR connector and one female XLR connector.
- for unbalanced outputs, use coax cables fitted with RCA Phono plugs.

Connect the power cable supplied to the power inlet on the DAC rear panel, plug the other end into a convenient power outlet.



Please do not use an excessively heavy or inflexible power cable as this may damage the power inlet connector.

Press the **Power** button and wait about 30 seconds while the DAC configures itself. The display will show in sequence: **Scarlatti DAC** and **No Input**.

If the unit is likely to be set in an unfamiliar state, you can reset it as follows:

Press the **Menu** button, press the **→** button once, then the **Menu** button again to display the **DAC Settings** menu. Press the **←** button a few times to highlight the **Factory Reset** menu page. Press the **Menu** button and wait a few seconds while the unit resets itself.

STEP 1 – Selecting a Digital Input

Switch on the source equipment. If appropriate, load a disk / tape and set the machine in PLAY mode to ensure it is generating a digital audio data stream.

Choose one of the following sections:

Connecting to an IEEE 1394 DSD Source

- Check that your source equipment (probably a *dCS* SACD Transport or a *dCS* Upsampler) is capable of DSD operation over a 1394 link, generates a 44.1kHz word clock and is set up correctly.
- Connect **one** 1394 output on your source equipment to **one** **1394** input on the *DAC* rear panel. Also connect the word clock output on your source equipment to the **Wordclock In** connector on the *DAC* rear panel.
- Press the **Input** button repeatedly until **1394** or the source name is displayed (probably **dCS STT**).

The unit may display **Wait ...**, **No-WCik** or **Search...**, messages if the unit is still settling, then in sequence (**Cik OK**), **Locking**, **DSD**. If you have more than one source connected to the 1394 bus, it may be necessary to use the **Input** button to select the 1394 source you want to listen to.

Connecting to a Single AES or SPDIF source

Most source equipment (such as CD transports, DVD players) is fitted with a single wire digital output, usually on an RCA phono connector.

- Connect your source equipment to the matching input on the *DAC* rear panel using a suitable cable.
- Press the **Input** button repeatedly until your chosen input is displayed. This will be either **AES1**, **AES2**, **RCA1**, **RCA2**, **Toslink** or **BNC**.

The *DAC* will lock to the source, displaying in sequence (**44.1k**), **Locking**, **16/44.1** for example, if the source is a CD player.

Connecting to a Dual AES Source

Check that your source equipment is capable of Dual AES operation.

- Connect the AES1 (or AES A) output on your source equipment to the **AES1** input on the *DAC* rear panel and the AES2 (or AES B) output to the **AES2** input, using two XLR cables. Ensure the cables are not swapped.
- Press the **Input** button repeatedly until both **AES1** AND **AES2** input indicators are lit.

The *DAC* will lock to the source, displaying in sequence (**96k**), **Locking**, **24/192** for example, if the source is 24 bit data at 192kS/s.

Connecting to an SDIF/DSD Source

Check that your source equipment is capable of SDIF PCM or DSD operation.



SDIF (Sony Digital InterFace) is not the same as SPDIF (Sony/Philips Digital InterFace) and the two are not compatible. Please ensure that you connect to the correct BNC socket.

- Connect the CH1 output on your source equipment to the **CH1** input on the *DAC* rear panel and the CH2 output to the **CH2** input, using two BNC cables. Connect the word clock output on your source equipment to the **WCik In** connector on the *DAC* rear panel. Ensure the cables are not swapped.
- Press the **Input** button repeatedly until **SDIF** appears on the display.

PCM or DSD mode is automatically detected. The *DAC* will lock to the source, displaying in sequence (**44.1k**), **Locking**, **16/44.1** for example, if the source is a CD player. **DSD** is displayed if the source is DSD.

Proceed to Step 2.

STEP 2 – Connecting the Analogue Outputs

Choose one of the following two sections:

Using a preamplifier

- Set the preamplifier volume control to minimum.
- Connect either the balanced (XLR connectors) or unbalanced (RCA phono connectors) outputs on *DAC* rear panel to matching line level inputs on your preamplifier (probably labelled CD or AUX).
- Turn the *DAC* rotary control clockwise to set the **Volume** to maximum (**-0.0dB** on the display).

Slowly increase the preamplifier volume until the music is at the right level.

Using a power amplifier directly

- Turn the *DAC* rotary control counter-clockwise to set the **Volume** to minimum (**-60.0dB** on the display).
- Connect either the balanced (XLR connectors) or unbalanced (RCA phono connectors) outputs on the *DAC* rear panel to matching inputs on your power amplifier. Switch on the power amplifier.
- Turn the *DAC* rotary control slowly clockwise until the music is at the right level.

Play a disc, you should have audio.

Proceed to Step 3.

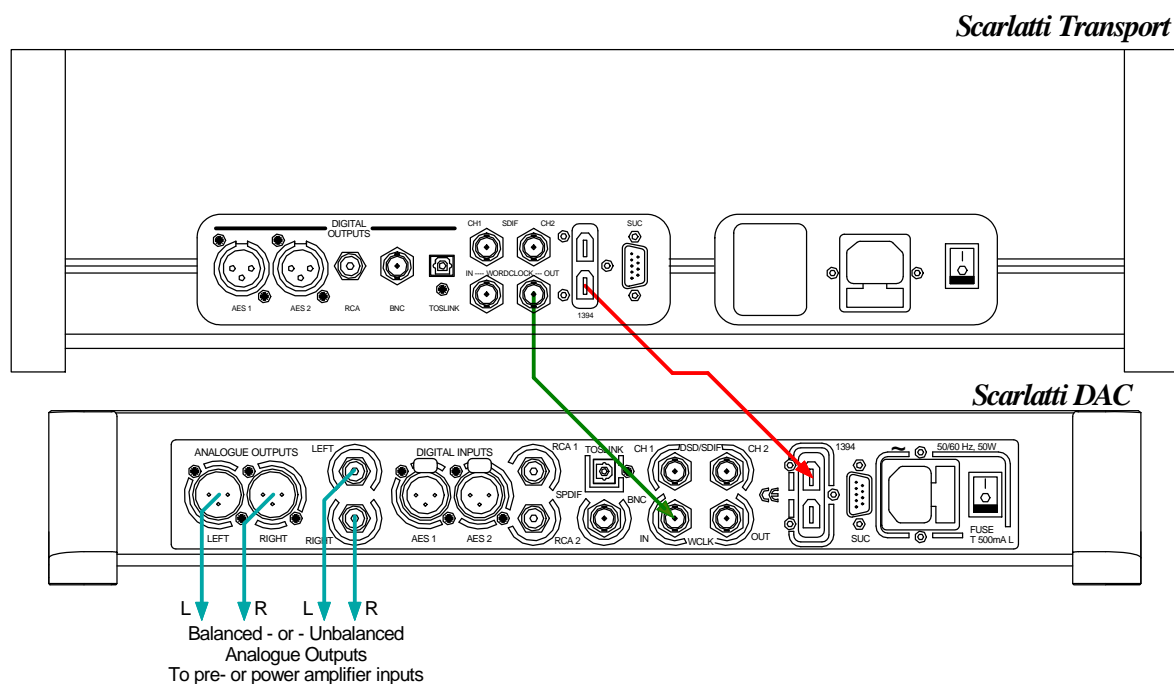


Figure 1 – Basic setup

STEP 3 - Setting the Output Level

If the preamplifier volume setting for a comfortable listening level is too high or too low, you may need to change the **Output Level** setting.



Setting the **Output Level** to **6V** can cause some preamplifiers to overload and distort. For this reason, we recommend the **2V** setting if a preamplifier is used.

Similarly, if you are driving a power amplifier directly and the **DAC Volume** setting for a comfortable listening level is higher than **-10.0** or lower than **-20.0** try changing the **Output Level** setting.

- Press the **Menu** button, press the **→** button once, then the **Menu** button again to display the **DAC Settings** menu. Press the **→** button a few times to highlight the **Analogue Out** menu page. Press the **Menu** button to change the setting.

If the setting is **2V** and the **Volume** is still set well below **-20.0**, try some passive attenuators.

STEP 4 – Choosing a Sync Mode

So far, the system has been set up to lock to the clock generated by the source. If you are using the 1394 or SDIF interfaces, the DAC will lock to a word clock generated by the source. This is the simplest arrangement, but it does not give the best sonic performance due to clock jitter.

If your source equipment has a word clock input, you may be able to reduce the jitter in your system by either setting the DAC to Master Mode or locking the system to a Master Clock. If not, you can miss out this section.

Using the DAC in Master Mode



You can use Master Mode **ONLY** if your source equipment can lock to a 44.1kHz word clock and the DAC is receiving data at 44.1, 88.2, 176.4 or 352.8kS/s or DSD. If the system is not set up correctly and fails to lock, you will hear clicks / noises from the speakers and the signal may be distorted.

Connect the equipment as shown, taking special care to connect the clock cable correctly.

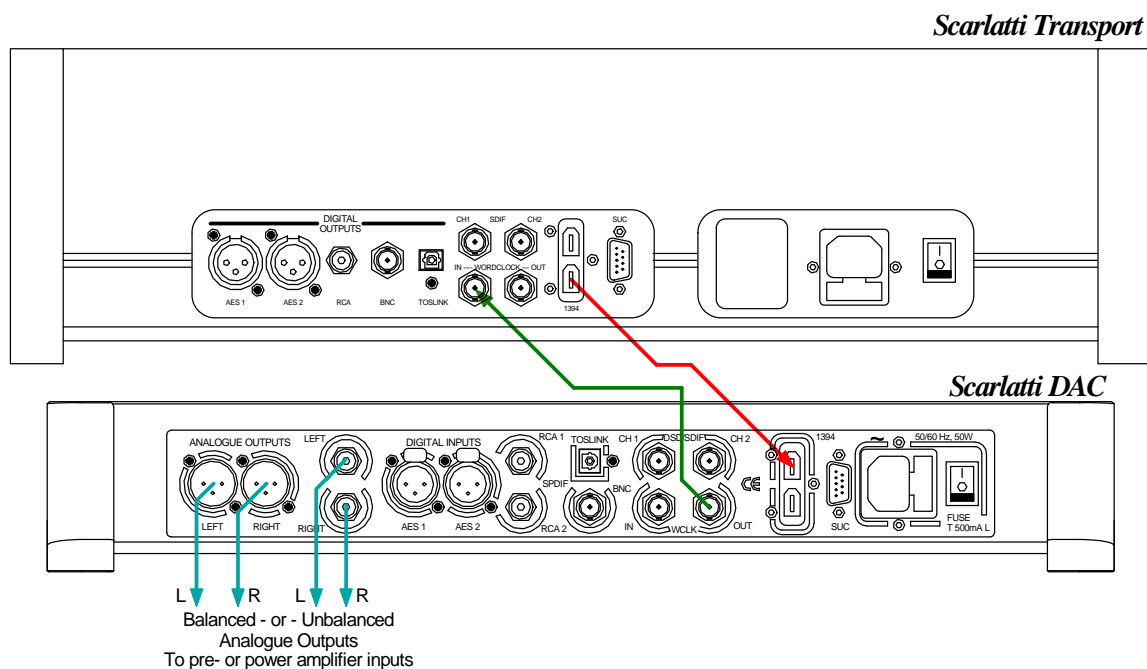


Figure 2 – Using the *Scarlatti* Transport with the DAC in Master Mode

- Connect the **DAC's Word Clock Out** connector to the word clock input on the source equipment. *dCS* transports slave automatically.
- Press the **Sync** button to set the DAC to **Master** mode and allow the system to re-lock.

Using a Master Clock

The performance can be improved further by adding a *Scarlatti* Clock to the system as shown below.

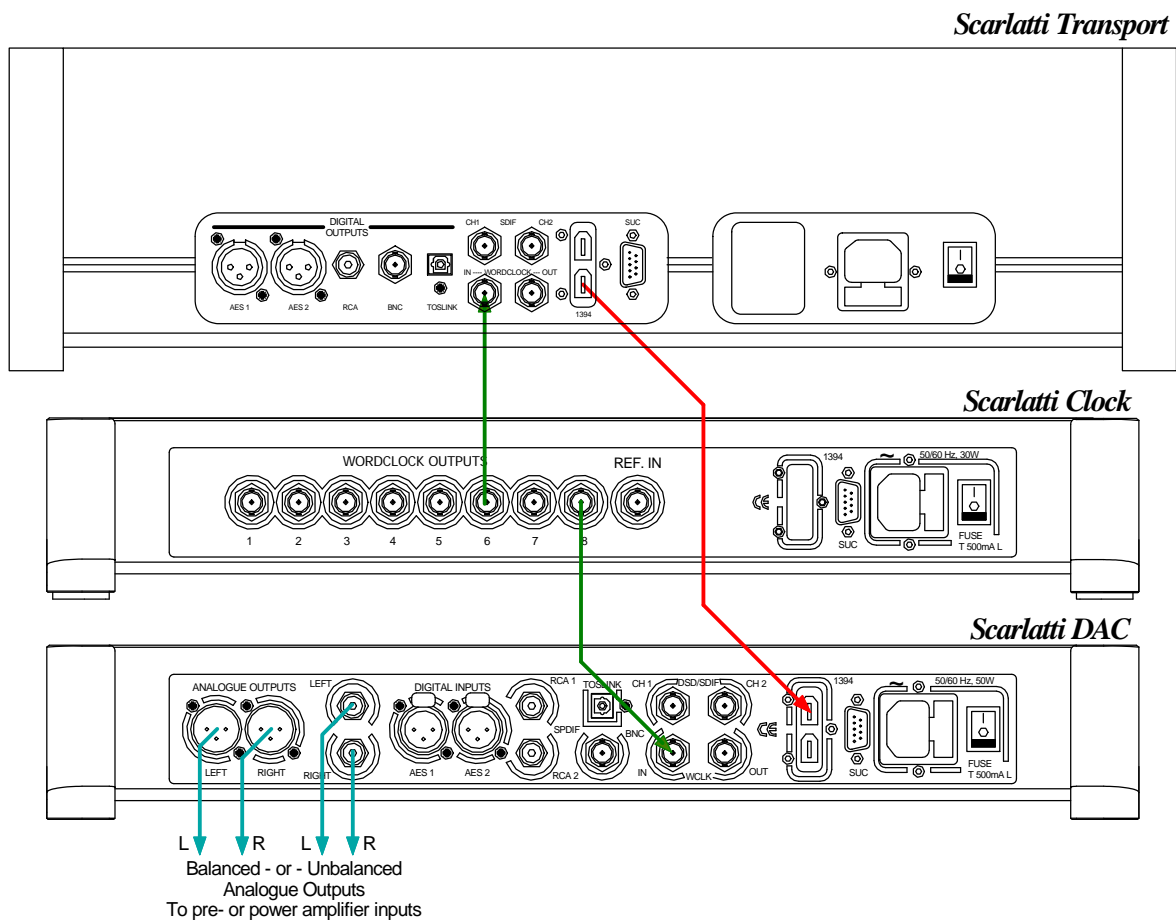


Figure 3 – Using the *Scarlatti* Transport, DAC and Clock together

- Connect one of the Clock's **Word Clock Outputs** to the **Word Clock Inputs** of the Transport and DAC.
- Use the Clock's **Frequency** button to set the outputs to **44.1kHz**, to suit the CD Transport.
- Use the DAC's **Sync** button to set the sync mode to **WClock**.

The system will re-lock and un-mute.

Please consult the *Scarlatti* Clock manual for more information on using a Master Clock.

Please consult the Upsampler manual if your system includes a *Scarlatti* Upsampler.

The next step is crucial – sit back and enjoy the music.

Using the *Scarlatti* DAC with a *Vivaldi* Transport

The *Vivaldi* series uses Dual AES to transmit DSD data between units, it has no 1394 interface.

Scarlatti DAC v1.30 allows the DAC to accept *dCS*-encrypted Dual AES data, so that *Scarlatti* system owners can upgrade to a *Vivaldi* Transport and still play SACDs in native DSD format, upsample CD data to DSD or DXD format. To do this:

- Connect the Transport's **AES 1 & 2** outputs to the DAC's **AES1 & 2** inputs respectively.
- Connect the DAC's **WCLK Output** to the Transport's **Word Clock Input**.
- Set the DAC's input to **Dual AES**.
- Use the DAC's **Sync** button to set the sync mode to **Master**.

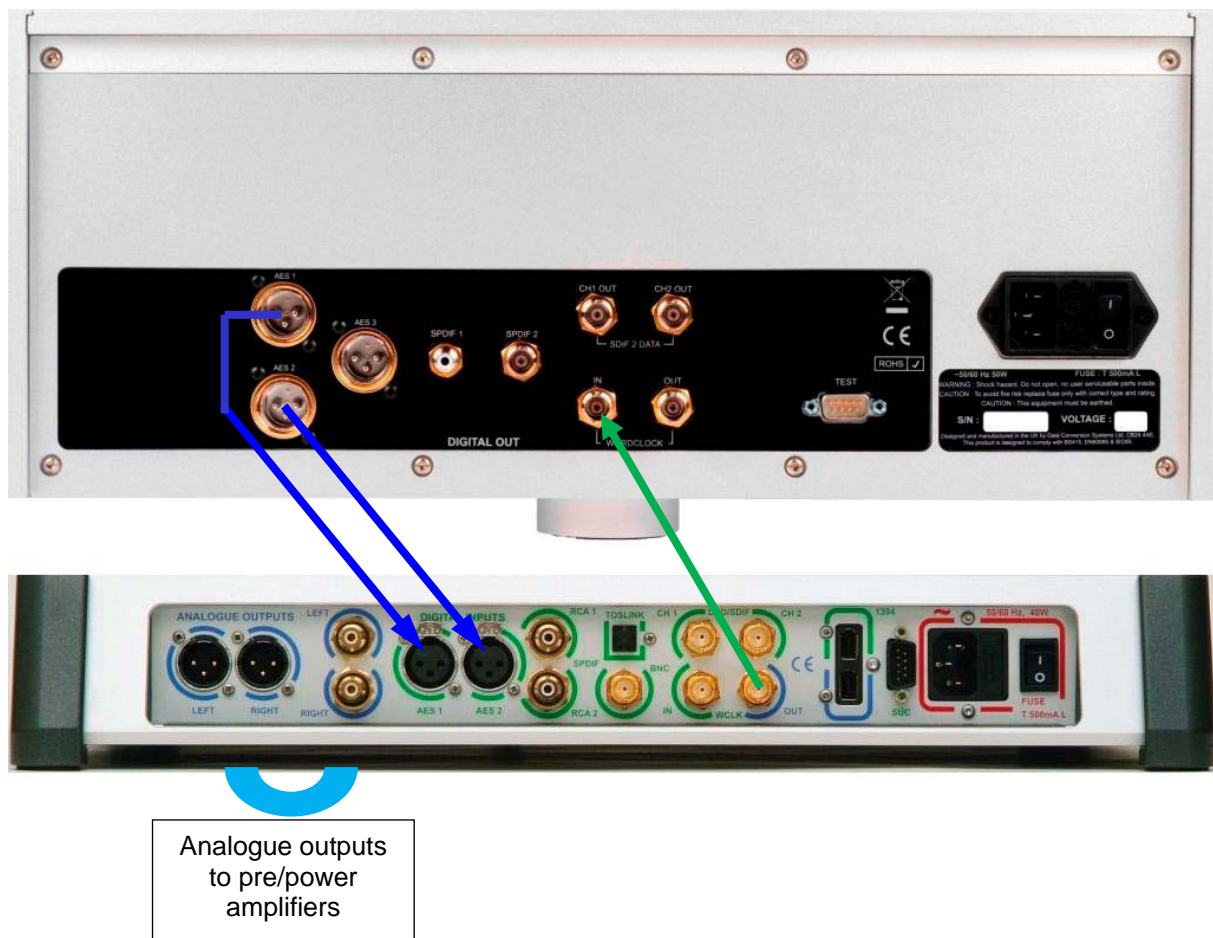


Figure 4 – Using the *Vivaldi* Transport with a *Scarlatti* DAC in Master Mode

The *Vivaldi* Transport's **Dual AES** menu page selects **DSD** or **DXD** upsampling for CD data. You can add a Master Clock (see page 11) or an Upsampler to the system if you wish.

FRONT PANEL

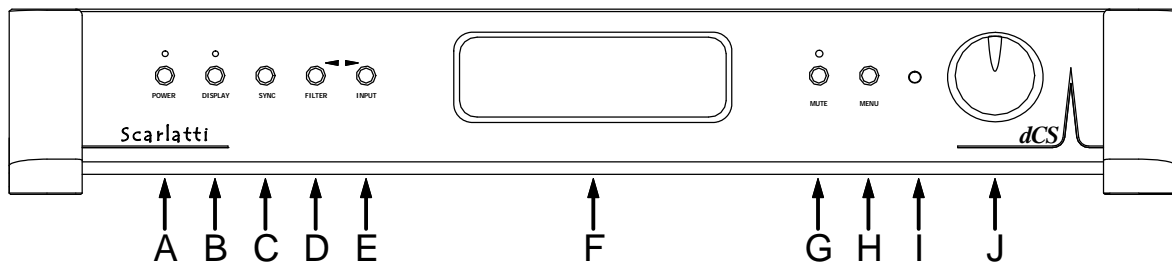


Figure 5 – Front panel

POWER Button

To switch on, ensure the rear panel switch is set to **I** and press the **POWER** button (A) on the front panel once. Note that the unit cannot be turned on from the remote control.

To set the unit to sleep mode, press the **POWER** button once. The main display will turn off, the LED over the button will light, and the analogue outputs will mute, but the unit will remain close to running temperature. Press again to return to normal operation.

To switch off, hold down the **POWER** button for about 5 seconds until **Power Down** appears on the display, then release it.

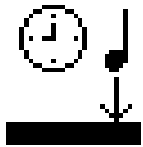
When the menu is open, press the **POWER** button to close the menu.

DISPLAY Button

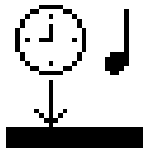
Press the **DISPLAY** button (B) to turn the main display off and on. When the display is off, the LED over the button will light to warn you that the unit is running.

SYNC Button

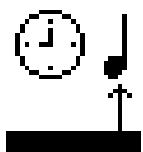
Use the **SYNC** button (C) to set the sync mode for the digital input that is currently selected. The DAC remembers the Sync setting separately for each digital input. The options are:



Audio – The DAC extracts the clock from the selected digital input and locks to it. Use this setting for systems that do not have a *Scarlatti* Transport. This arrangement does NOT give the best jitter performance or the best sound quality.



WClk – The DAC locks to an external clock connected to the **Word Clock Input**. This is the normal setting if you are using a *Scarlatti* Transport or a *Scarlatti* Clock or both. For the very best sound quality, use this setting with the full stack.



Master – The DAC uses its own stable 44.1kHz clock and outputs the same clock on the **WClk Output**, so that the source can lock to the DAC. For correct master mode operation, the source equipment MUST be locked to the DAC. This is the recommended setting if you have a *Scarlatti* Transport, but not a *Scarlatti* Clock.



You will not be able to set an input to Master mode unless the DAC is locked to 44.1, 88.2, 176.4 or 352.8kS/s or DSD data on that input.

The SDIF input cannot be set to Master mode unless the DAC is locked to a 44.1 or 88.2kHz word clock connected to the **Word Clock Input**.

FILTER Button

Use the **FILTER** button (D) to select your preferred filter. This is a personal preference and is best done by ear. The choice of filter may depend on the type of music you are listening to.

In PCM mode, the first 4 filters give different trade-offs between the Nyquist image rejection and the phase response. Filter 1 has the best rejection of (unwanted) Nyquist images and the sharpest roll-off, resulting in the poorest transient response of the four. Filters 2, 3 and 4 have progressively more relaxed image rejection and progressively better transient response. **Filter 2** is often preferred for orchestral music, while **Filter 3** and **Filter 4** are often used for rock music.

If the source data rate is 176.4 or 192kS/s, two extra filters are available. **Filter 5** has a Gaussian response and **Filter 6** is an asymmetrical type which features almost no pre-ringing.

There are 2 extra filters for 44.1kS/s operation. **Filter 5** is an asymmetrical design with non-linear phase and no pre-ringing. **Filter 6** is a new sharp filter which has linear phase and pre-ringing. Try them and decide for yourself which you prefer.

DSD mode has 4 filters, but these progressively reduce the out-of-band noise level (which is inherent in the 1-bit nature of DSD). **Filter 1** is the usual setting – it gives the widest bandwidth (about 70kHz) and the highest level of out-of-band noise. If your system sounds harsh, try **Filter 2** or **Filter 3**. These progressively reduce the out-of-band noise level at the cost of some bandwidth. **Filter 4** is primarily intended for troubleshooting, not listening, as it has an early cut-off, designed to minimise the out-of-band noise.

The DAC remembers the last filter set for every sample rate.

When the menu is open, the **FILTER** button changes to the ← button, used to page backwards through the menu.

INPUT button

Press the **INPUT** button (E) repeatedly to cycle through the available digital inputs. The sequence is: **AES1**, **AES2**, **DUAL AES** (this disappears from the list if it is turned off in the **Dual AES** menu), **RCA1**, **RCA2**, **Toslink**, **BNC**, **SDIF-2**, **1394** (the name of an active source will be displayed, e.g. **dCS STT** or **dCS SUP**).

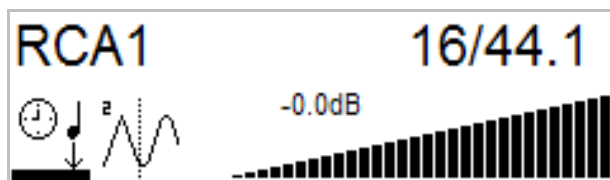
When the menu is open, the **INPUT** button changes to the → button, used to page forwards through the menu.

Display

In normal use, the display (F) is split into 4 areas:

The top left corner usually shows the selected input (e.g. **RCA1**) or 1394 source name (e.g. **dCS STT**). While you are changing settings, the current setting may be displayed here briefly.

The top right corner shows the locking status, the data format. Some examples are **No Input**, **Locking**, **(Clk OK)**, **16/44.1** (the number of active bits and the sample rate), **24/192** or **DSD**. Between tracks on a CD, the number of active bits is zero, so **0/44.1** may be displayed.



The icons at the bottom left corner are the **Sync** mode (slave to **Audio** is shown) and the selected **Filter** (**Filter 2** is shown). The **Phase** icon appears here if set to inverted.

The “staircase” icon gives a visual indication of the **Volume** or **Balance** setting, with the setting shown in text as well (**-0.0dB** is shown).

Details of the menu displays are shown in the Menu section on page 18.

MUTE Button

Use the **MUTE** button (G) to mute the analogue outputs. When the unit is muted (either manually or when not locked to a source) the LED over the button lights.

MENU Button

Press the **MENU** button (H) to open the menu, select menu pages and change settings. See the Menu section on page 18 for information on using the menu features.

Remote Control Receiver

Aim the remote control handset towards the receiver (I) for best sensitivity.

Rotary Control

The Rotary Control (J) usually controls the **Volume** setting.

When the **Balance** menu page is open, it adjusts the channel **Balance** instead.

If the menu is open, the Rotary Control pages forward and backwards through the menu.

REAR PANEL

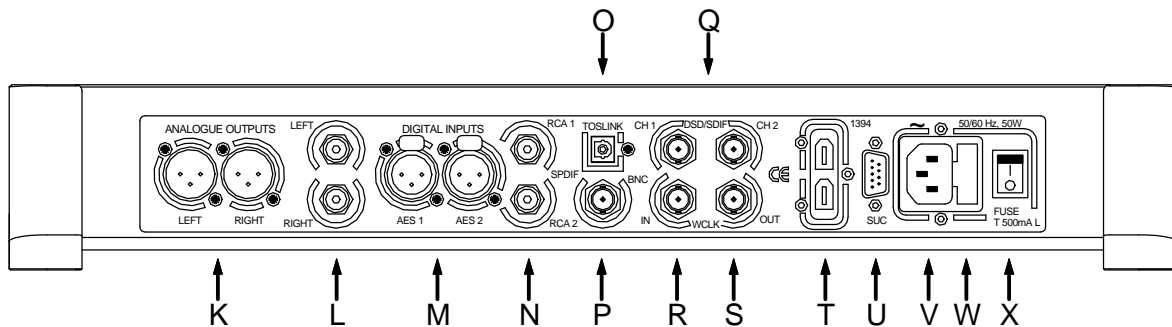


Figure 6 – Rear panel

Analogue Outputs

The unit features separately buffered **Balanced Outputs** (K) on XLR connectors and **Unbalanced Outputs** (L) on RCA connectors. The **Balanced Outputs** should be connected to true balanced inputs only. If your preamplifier / power amplifier has unbalanced inputs (even if the inputs are on XLR connectors), please use the **Unbalanced Outputs** instead.

AES Digital Inputs

The **AES1** and **AES2** inputs (M) can be used individually at sample rates up to 192kS/s or DSD in DoP format.

If the **Dual AES** menu page is set to **On** or **Auto**, they can be used together as a **Dual AES** pair at 88.2, 96, 176.4, 192, 352.8 or 384kS/s or DSD in DoP format or with *dCS*-encrypted DSD. For Dual AES mode to work correctly, the source must actually generate Dual AES data, not just single AES on 2 connectors!

SPDIF Digital Inputs

The unit features 3 co-ax SPDIF inputs, labelled **RCA1**, **RCA2** (N) and **BNC** (P), as well as an optical SPDIF input on a **Toslink** connector (O). Pull out the dust cover before using the **Toslink** input. These inputs will accept sample rates up to 192kS/s or DSD in DoP format, but note that operation of the **Toslink** input is not guaranteed above 96kS/s.

DSD/SDIF Digital Interface

The **DSD/SDIF** interface will accept either SDIF-2 PCM data at sample rates up to 96kS/s, or SDIF-2 DSD data. The unit automatically detects the data format and sets the correct mode. The interface consists of two data inputs labelled **CH1** and **CH2** (Q). Operation in SDIF mode requires that a word clock from the source is connected to the **WCLK In** connector (R).

Word Clock Input / Output

The **WCLK In** connector (R) will accept standard word clock from the source equipment or a master clock at 32, 44.1, 48, 88.2, 96, 176.4 or 192kHz. The clock frequency **MUST** be an exact multiple of the data rate, otherwise the system will not lock. Use the **Sync** button to lock the DAC to the external word clock. Note that the source **MUST** be locked to the same clock, otherwise the system will not be locked and periodic clicks will be heard on the outputs.

When the unit is set to Master mode, the DAC uses its internal clock instead of locking to the data or **WCLK In**put, and the **WCLK Out** connector (S) carries a 44.1kHz word clock. This **MUST** be connected to the source equipment, so that the system can lock.

Word clock is used for synchronisation only, it does not carry digital data.

1394 Interface

The IEEE 1394 interface (T) will accept encrypted DSD data from the *Scarlatti* Transport or Upsampler, as well as from other *dCS* transports and upsamplers that are loaded with current software. 1394 interfaces are great for carrying lots of digital data - but are very poor at transmitting a stable clock, so a separate 44.1 or 88.2kHz Word Clock connection is required. The two 1394 ports are identical – they can be used as data inputs or as a loop-through. Note that the 1394 bus must not be connected in a loop as this will prevent the system initialising.



Due to proprietary encryption of the DSD data, the 1394 interface must be assumed to be incompatible with other manufacturer's equipment.

SUC Connector

The **SUC** connector (U) is an RS232 interface, primarily used to remotely control the unit during automated testing. Please contact *dCS* for advice on using this interface with a household automation system. Note that we recommend using infra-red remote control instead.

Mains inlet

Power is connected via a standard IEC320 connector (V), protected by a fuse (W) and isolated by a 2-pole power switch (X).

THE MENU

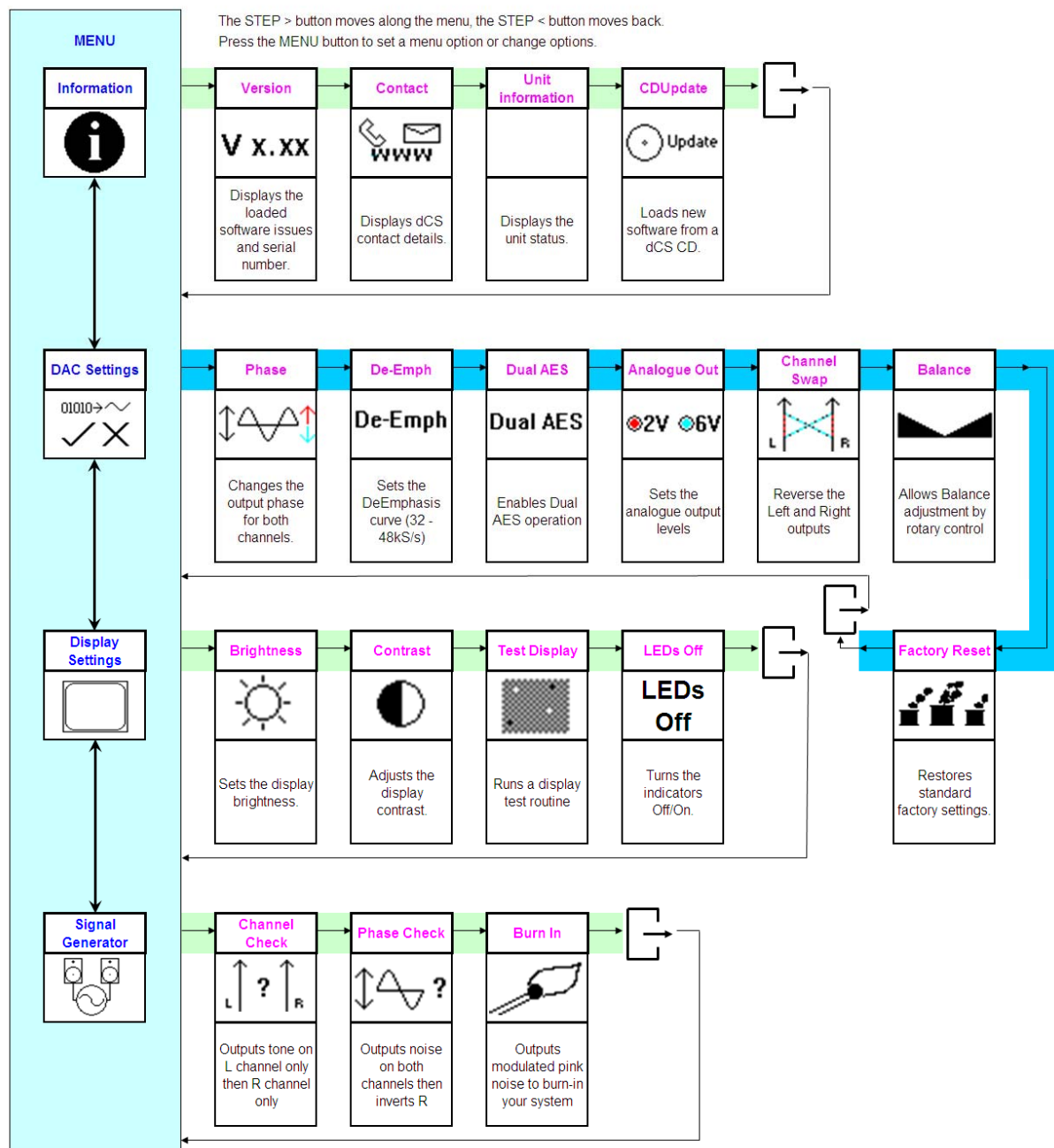


Figure 7 – The menu sequence

Using the Menu

The menu gives the user access to a range of additional features. It also allows new features and performance enhancements to be added at a later date by software upgrades.

The menu is controlled by four buttons.

- Press the **MENU** button to open the menu or select a setting.
- Press the **→** button to page forward through the menu.
- Press the **←** button pages backward through the menu.
- Press the **POWER** button to close the menu or just wait 5 seconds.

Alternatively, use the remote control to access the menu.

Use the Menu Guide sheet to help you find the right menu.

Each unit in the range has either three or four top-level menu pages:



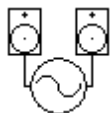
The **INFORMATION** menu gives software issues, serial number, contact details and unit set-up details.



Each model has a different **SETTINGS** menu, which allows you to set some features that are not directly accessible from the front panel.



The **DISPLAY SETTINGS** menu is used to adjust and test the display.



The **SIGNAL GENERATOR** menu (on the Transport and DAC only) contains test and set-up routines.

Use the **→** button to move the highlight to the menu you want, then press the **MENU** button to select it. The next menu level down is displayed. Use the **→** button to move the highlight to the menu page you want, then press the **MENU** button to display the information or change the setting.



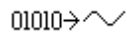
Select the exit icon to go back to the previous menu level.

INFORMATION Menu

V X.XX The **Version** page displays the software versions loaded in the unit and the full serial number. Please have this information ready if you contact your dealer for any reason.



The **Contact** page displays *dCS* web-site URL, email address, telephone and fax numbers. If you have any difficulty, please contact your dealer for help first.



The **DAC Information** page displays the unit status:

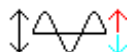
- No input / Input selected to
- 1394 nodes connected:
- Sync to Audio / WClk / Master
- Unlocked / Locked at (either word clock frequency or data sample rate)
- Input is PCM / DSD
- No De-Emphasis / De-Emphasis is
- Filter
- Phase Normal / Inverted.
- Internal temperature°C

Use the rotary control to scroll down the list and press the **MENU** button to exit.



The **CD Update** feature allows you to load new software into your system from either a *Scarlatti* transport or any STANDARD CD player or transport. Note that some non-Red-Book CD transports change the digital data and cannot be used to download new software. Please follow the instructions supplied with the update CD carefully.

DAC SETTINGS Menu



Changes the **Phase** of all analogue outputs from normal (**↑**) to phase reversed (**↓**).

De-Emph

Apply **De-Emphasis** to equalise the frequency response of pre-emphasised data. **De-Emphasis** is available for sample rates of 32, 44.1 or 48kS/s only. The settings are:

- **Auto** – The unit detects the emphasis flag of the digital data and automatically applies the correct de-emphasis. This is the usual setting. Note that if the emphasis flag in the data is wrong, you will need to manually select the correct setting.
- **Off** – No de-emphasis is applied.
- **50/15** – Applies the 50/15us de-emphasis curve and ignores the flag. This was sometimes used on early CDs.
- **CCITT** – Applies the CCITT J17 de-emphasis curve and ignores the flag. This is used for DAB radio transmissions.

Dual AES

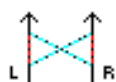
This setting allows the two AES inputs to be used together to accept PCM data at high sample rates (88.2, 96, 176.4, 192, 352.8 or 384kS/s or DSD) from a Dual AES source. Note that a source with two AES outputs might not actually produce Dual AES data!

The settings are:

- **Auto** – The unit detects the format flags in the digital data and automatically sets the **AES1** & **2** inputs to Single AES or Dual AES mode as required. This is the usual setting. Note that if the format flag in the data is wrong, you will need to manually select the correct setting.
- **Off** – Dual AES mode is disabled.
- **On** – Dual AES mode can be manually selected using the **INPUT** button or the remote control.



Sets the **Output Level** to **2V** rms or **6V** rms at full scale. When you set up your system, choose the **Output Level** setting that gives a comfortable listening level with the **Volume** set between **-10dB** and **-20dB**. The difference between the two settings is about 10dB.



Channel Swap - swaps the channels to correct a connection error. Correct the error and change back to normal (**↑↑**).



In **Balance** mode, the operation of the rotary control changes to adjust the **Balance**. Each channel can be varied from **0dB** to **-6dB** in 0.1dB steps, then mute ($-\infty$) below that. **Balance** adjustment ends after this menu page closes or the **MENU** button is pressed. Most owners use the remote control to adjust the balance.

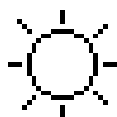


The **Factory Reset** menu page resets the unit to standard settings. These are:

Volume to **-30dB**.
Balance to centre.
Input to **AES1**.
Sync to **Audio** Input except:
1394 & **SDIF Sync** to **WCik**.
De-Emphasis to **Auto**.
Filter to 1.

Phase to **Normal**.
Dual AES to **Auto**.
Output Level to **6V**.
Channel swap to **Normal**.
Display to **On**.
Brightness to maximum.
Contrast to 60%.

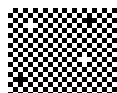
DISPLAY SETTINGS Menu



Set the display **Brightness** to a comfortable level using the rotary control.



Set the display **Contrast** to a **comfortable** level using the rotary control.

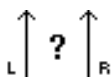


Runs a **Display Test** routine, which flashes all the indicators and the main display.

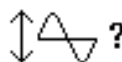
LEDs Off

This page is usually set to **LEDs On**. Set it to **LEDs Off** to turn off the indicators if you find them distracting.

SIGNAL GENERATOR Menu



Runs a **Channel Check** routine to test the system for swapped channels. A tone is output on the left channel only while the display shows **Left**, then the right channel only while the display shows **Right**. If these are the wrong way around, the left and right channels are swapped somewhere in your system.



Runs a **Phase Check** routine to test the system for phase reversals. Noise is output in phase on both channels while the display shows **In Phase**. Press the **MENU** button, the noise on the right channel is inverted while the display shows **Out of Phase**. The first burst should produce a central image, the second burst should not. If these are the wrong way around, one channel in your system is phase inverted. Press the **←** or **→** button to stop the test and return to the menu.



Runs a **Burn-In** procedure to condition your system. The unit displays **Burn In Caution! Loud**, then outputs modulated pink noise that ramps up slowly in level. Press the **MENU** or **← →** buttons to stop the procedure. Please ensure that the volume level is reasonable, as careless use can damage your amplifiers and loudspeakers. dCS will not be liable for such damage.

SPECIFICATION

Converter type	5-bit, 2.822 or 3.07MS/s oversampled, <i>dCS Ring DAC</i> topology.		
Digital inputs	IEEE 1394 interface on 2x 6-way connectors. The interface accepts <i>dCS</i> -encrypted DSD (1 bit data at 2.822MS/s) from up to 4 sources.		
	2x AES/EBU on 3-pin female XLR connectors. Each will accept up to 24 bit PCM at 32 - 192kS/s or DSD in DoP format, OR as a Dual AES pair at 88.2, 96, 176.4, 192, 352.8 or 384kS/s or <i>dCS</i> -encrypted DSD or DoP format.		
	3x SPDIF on 2x RCA Phono and 1x BNC connectors. Each will accept up to 24 bit PCM at 32 - 192kS/s or DSD in DoP format.		
	1x SPDIF optical on a Toslink connector, accepts 24 bit PCM at 32 -96kS/s.		
	1x SDIF-2 interface on 2x BNC connectors, will accept up to 24 bit PCM at 32 - 96kS/s or SDIF-2 DSD (auto-selected). If the unit is not in Master mode, this interface requires a compatible word clock input, locked to the data rate.		
Clocking	Word Clock Input on 1x BNC connector, accepts standard word clock at 32, 44.1, 48, 88.2, 96, 176.4 or 192kHz. The data rate can be the same as the clock rate or an exact multiple (0.25x, 0.5x, 1x, 2x, 4x) of the clock rate. Sensitive to TTL levels.		
	Word Clock Output on 1x BNC connector. In Master mode, a TTL-compatible 44.1kHz word clock appears on this output, not temperature compensated.		
Frequency response (set to Filter 1)	Fs = 32kS/s	+0.1/-0.5dB, 10Hz to 15kHz	
	Fs = 44.1 or 48kS/s	+/-0.1dB, 10Hz to 20kHz	
	Fs = 88.2 or 96kS/s	+/-0.1dB, 10Hz to 20kHz	-3dB @ >38kHz
	Fs = 176.4 or 192kS/s	+/-0.1dB, 10Hz to 20kHz	-3dB @ >67kHz
	Fs = 352.8 or 384kS/s	+/-0.1dB, 10Hz to 20kHz	-3dB @ >100kHz
	DSD	+/-0.1dB, 10Hz to 20kHz	-3dB @ >66kHz
Residual noise	Better than -110dB0, 20Hz - 20kHz unweighted.		
Spurious responses	Better than -100dB0, 20Hz - 20kHz.		
L-R crosstalk	Better than -80dB, 20Hz - 20kHz.		
Output levels	2V rms or 6V rms on all outputs for a full-scale input, set in the menu.		
Balanced outputs	1 stereo pair on 2x 3-pin male XLR connectors (pin 2 = hot, pin 3 = cold). These outputs are electronically balanced and floating, the signal balance ratio at 1kHz is better than 40dB. Output impedance is 3Ω, maximum load is 600Ω (a 10kΩ - 100kΩ load is recommended).		
Unbalanced outputs	1 stereo pair on 2x RCA Phono connectors. Output impedance is 52Ω, maximum load is 600Ω (a 10kΩ - 100kΩ load is recommended).		
Size and weight	465mm (18.3") long x 405mm (16.0") deep x 75mm (3.0") high. Allow extra depth for cable connectors. Allow space for air flow around the unit. Weight: 11.3kg (24.9lbs).		
Power requirements	Internally set to either 100, 115/120, 220 or 230/240V AC, 49 – 62Hz. Power consumption: 22W typical, 30W maximum.		

These specifications are subject to change without notice.

MAINTENANCE AND SUPPORT

Service and Maintenance

dCS audio products are designed not to need regular maintenance, and contain no user serviceable parts apart from the mains fuse. If your unit is damaged in any way, please contact your dealer.

Replacing a Blown Mains Fuse

There is a mains fuse below the power inlet, accessible from the outside of the unit. If the fuse blows, it may be changed by the user. The current consumption of the unit is very low, so it only blows if power surges occur, or there is a fault in the unit. Usually power surges cause no other damage, but if the fuse blows repeatedly on replacement, some other damage will have been done and the unit must be returned to dCS for repair.

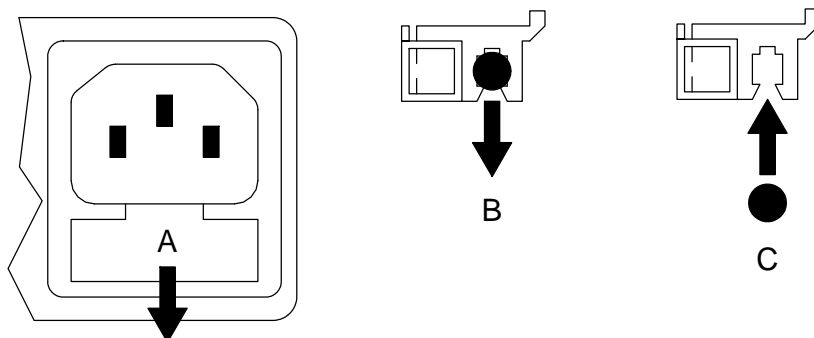
Fuse type: 20 x 5mm T0.5 amp L fuse



If the fuse should fail, it is essential that it is replaced with one of the same type and rating. Failure to do so could result in damage to the unit, risk of fire or electric shock and will invalidate the warranty.

Referring to the diagram below, remove the power cable, use a small flat bladed screwdriver to pry up the tab on the fuse carrier (A) and pull it out. Push the blown fuse out of the clip in the carrier (B) and dispose of it. Fit a new fuse in the clip (C) and push the carrier back into the unit so that it clicks home.

Spare fuses are provided with the unit.



Cleaning the case

The front and back panels of your dCS equipment are machined from very high grade aluminium. Great care has been taken to create the finish of the aluminium throughout the engineering process from the raw solid material to the finished piece.

To remove loose dust or finger marks from the case, we recommend that you use a clean, dry, lint-free cloth.

To restore the finish on the front and back panels, we recommend applying small quantities of a lanolin based cleaner, using a clean, dry, lint-free cloth and then wiping off. Do not allow lanolin to collect around the buttons.

Small amounts of glass cleaner containing ammonia may be used to clean other surfaces, but avoid spraying onto the connector contacts.

Limited Warranty

General

dCS warrants this product against defects in materials and workmanship for a period of 3 years from the date the unit was originally shipped from dCS. During the warranty period, dCS will repair or, at our absolute discretion, replace a faulty product. Warranty repairs must only be carried out by dCS or our authorised service agents. Please contact your dealer if your unit requires service.

Your dealer should have completed on your behalf an **Owner Registration** form at the time of sale and returned it to dCS. On receipt of the Owner Registration form, dCS will add your contact details to our customer database. dCS will use this information for warranty purposes only, we will not contact you directly for reasons relating to sales and marketing.

This warranty applies to the original owner.

Warranty Exclusions

The Warranty does not cover wear and tear.

The Warranty on this product will be void if:

- the product is misused in any way.
- any unauthorised modifications or repairs are carried out.
- the product is not used in accordance with the Operating Conditions stated in this manual.
- the product is serviced or repaired other than by dCS or our authorised service agents.
- the product is operated without a mains earth (or ground) connection.
- the unit is returned inadequately packed.

dCS reserve the right to apply a service charge if a product returned for warranty repair is found to be operating correctly, or if a product is returned without a returns number being issued.

This warranty covers parts and labour only, it does not cover shipping charges or tax/duty.

Our dealers or distributors are NOT authorised to extend the terms of this warranty, dCS cannot accept responsibility for any attempt to do so.

Products re-sold by dCS on a "used" basis may be subject to reduced warranty terms.

Obtaining Service

Should you encounter a problem, contact your authorised dCS dealer for advice, quoting the model, the full serial number, software version number, and giving a detailed description of the fault. Your dealer will advise you fully on actions that need to be taken. When returning a unit, the original packaging should be used to avoid transit damage. Replacement packaging sets may be purchased from dCS.

During the Warranty period, there will normally be no charge for parts or labour.

Operating Conditions

- The supply voltage must remain within +/-10% of the A.C. voltage specified on the back panel.
- The supply frequency must be in the range 49Hz to 62Hz.
- Ambient temperature range: 0°C (32°F) to 40°C (104°F), non-condensing.
- Do not install the unit near heat sources such as radiators, air ducts, power amplifiers or direct strong sunlight.
- If in doubt, the easy test is – the unit is happy to work anywhere a human is.

If you need more help

In the first instance, you should contact your dealer. If they cannot resolve the issue, contact your national distributor.

Manufactured by:

Data Conversion Systems Ltd.
Unit 1, Buckingway Business Park,
Anderson Road,
Swavesey,
Cambridgeshire. CB24 4AE
UK

www.dcsLtd.co.uk

This user manual may be downloaded free of charge from our web-site.

A bound copy of this manual may be ordered from *dCS*.

Software History

dCS products make extensive use of software configurable chips – FPGAs and DSPs. This gives us the ability to update our products to add extra features, update digital interface standards or make performance improvements by loading new software. Occasionally, a hardware upgrade may be necessary also to increase the “capacity” of the electronics, add extra connectors or extra front panel controls.

Please note that not all software updates make an earth-shattering change. You should have a clear idea of what you expect to gain before updating to the latest issue. We recommend that you keep your software up to date. Check the *dCS* web-site for the latest software updates.



Before requesting an update disc, please open the **Information > Version Information** menu page and check the **Control Version** to verify that an update is actually needed.

This manual is for *Scarlatti DAC* software version 1.3x.

Issue 1.00 - The first issue, with 1394 issue 3.00.

Issue 1.01 – Corrects a bug that causes some DACs to momentarily lose lock to the Word Clock Input.

Issue 1.10 – Adds Filter 5 and Filter 6 at 44.1kHz.

Adds an IR remote command for Sync to Word Clock.

Updates the EasyPlay system to accommodate an Upsampler on 1394 channel 1.

LEDs Off feature added.

Issue 1.20 – Adds single wire 176.4 & 192kS/s operation to AES, RCA, BNC and Word Clock inputs.

Issue 1.30 – AES, Dual AES, RCA & BNC inputs will accept DSD in DoP format.

The Dual AES input will accept *dCS*-encrypted DSD, DSD in DoP format, 24/352.8 or 24/384 (this adds compatibility with a *Vivaldi* Transport and Upsampler).