



2 Channel Microphone Preamplifier



**USER GUIDE** 



# Instructions for use and installation

#### Welcome

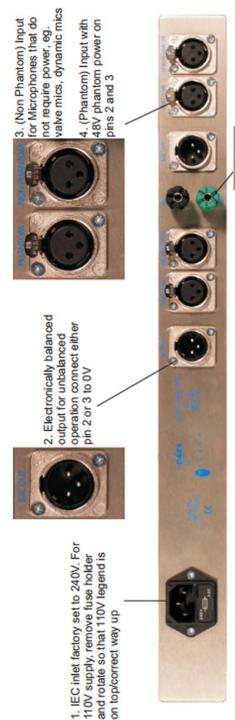
At DACS we are very pleased that you have chosen to purchase one of our products. We take pride in our work and are sure that this MicAmp2 will give you years of exemplary service. If you have any suggestions or comments about this product please call, fax, write or e-mail us with your thoughts. Thank you.

#### Introduction

The MicAmp2 is an ultra low noise microphone amplifier. It uses three super matched pairs of transistors for the main gain stage, giving low noise performance and very fine detail, even at very low input levels. The frequency response is flat down to below 1Hz at the bottom (DC for non phantom powered microphones) and does not start its tailored high frequency roll-off until above 45kHz. The CMRR is hand trimmed at mid and high frequencies to very high tolerances.

Each channel has an eight position self cleaning rotary switch to set gain in 6dB steps from 20dB to 62dBd. There is then a gain trim from 0dB to +10dB, and a three position bass roll off filter, giving a flat response and two cut-off points at a nominal 80Hz and 30Hz. The cut-off point and slope of the filters have been chosen to give a smooth response, while removing very low rumble (30Hz) and more mains type noise (80Hz). In addition they have no effect above their cut-off frequency. The balanced output is a fully floating servo type output, with hand trimmed balance set to +/-1% accuracy, delivering up to +22dB into 600R.

Working with the MicAmp2 is straightforward. Settings are easily repeatable due to the stepped gain and detailed scaling on the trim control. There is ultra fast protection on the input so that any fault condition immediately triggers the protection circuitry, preventing damage to the amplifier circuitry.



9. Unique tri segment LED meter - PPM ballistics variable intensity LED illumination - Green illuminates at -15dBM, Yellow at +5dBm and Red at +13 dBm

10. Ultra bright overload light illuminates when clipping occurs (overload ca +22dBm) and remains on for a few seconds

**Polarity** 

5. Audio 0V, isolated

from mains earth



7. Gain Trim allowing extra gain of 10dB scale to the hour and half hour for easy repeatability (6 o'clock = 0dB. 3 o'clock= +10dB)

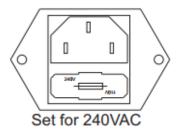
8.Bass roll off filter - flat and 30Hz or 80Hz the filter curve has been selected to minimise the effect of frequencies above cut off point

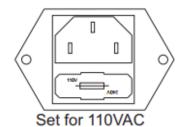
# Using the DACS MicAmp2

#### Installation

### 1 Connecting the Power

The unit will accept 240 VAC and 110 VAC mains supplies. The IEC inlet's fuse holder is used as a selector as shown in Fig 2. The factory setting is for 240 VAC. Figure 2





## 2 Outputs

The output from the DACS MicAmp2 is a fully floating electronically balanced output designed to drive down long  $600\Omega$  lines. It will run balanced or unbalanced, and can provide signal levels in excess of +22dBm when running balanced or unbalanced. To run unbalanced, connect either pin 2 or pin 3 to 0V. NB The output will not run unbalanced satisfactorily if one leg is left floating and not connected to 0V.

## 3 & 4 Inputs

The DACS MicAmp2 has two microphone inputs, one for phantom powered (48V) microphones and one for non phantom powered microphones. The non-phantom powered input by-passes the input capacitors, and avoids the signal degradation caused to low microphone voltages by switching the signal. The microphone amplifier stage has a minimum gain of 20dB. Before connecting microphones, ensure that the MicAmp2's gain is set to 20dB, and that any loudspeakers are turned down.

#### 5 Earthing

The signal  $\overline{0V}$  is separate from the mains earth. The mains earth is connected to the case of the device. If you wish to connect the signal earth to the mains earth or a technical earth, this may be done by running a wire from the earth tag. This connection should only be made by qualified personnel, if it is necessary. If in doubt please call us.

#### 6 & 7 Setting Gain

The gain for the MicAmpII is set using the stepped GAIN control in 6dB steps, and then trimmed between 0dB and +10dB using the continuously variable TRIM control. The scaling on the trim control allows accurate repetition of settings. The main lines may be thought of as hours (0dB is 6 o'clock, and +10dB is 3 o'clock), while the secondary lines are half hours. The settings may then be remembered or noted down as times, making it easy to repeat them accurately. Optimum gain settings vary, but see below for operating levels and overload indication.

#### 8 Roll Off

The bass roll of is designed to minimise the pick up of unwanted rumble, and to have little or no effect on the signal above the roll off frequency. The slope of 12dB / octave and cut off frequency were selected to minimise very low earth borne rumbles (30Hz cut-off) and more buzzy mains type noises (80Hz cut-off). The filter has no effect on frequencies above the cut-off, and causes no 'bumps' in the frequency response.

## 9 & 10 Metering

The signal level and condition is show by the unique tri-segment LED meter and the ultra bright overload LED. The segments of the LED meter illuminate increasingly brightly as the signal level increases, with the Green LED illuminating first to indicate that signal is present, the Yellow LED illuminates at +5dBm, achieving full brightness at +13dB when the Red LED begins to illuminate. The Overload LED (!) illuminates when the signal clips, and remains on for a few seconds.

#### 11 Phase Reversal switch

This gold plated contact switch reverses the signal phase when it is down. It has no other effect.

#### 12 Powerswitch

This switches the power on and off, and is illuminated when AC mains is present. This switch is connected before the fuse, so when it is illuminated this indicates that mains is present. If the unit does not work, the fuse may be blown. The unit comes with a spare fuse in the fuse carrier. Replacement fuses should be T1A.

# Specifications for MicAmp2

Distortion THD+N @ 1kHz <0.0016% @ 1kHz with +10dB output

and 60dB gain

Self Noise -128dB (DC-40kHz)

Cross talk <90dB

Frequency response Phantom Power Flat <2Hz to ca 45kHz, tailored roll off;

Non Phantom DC to ca 45kHz, tailored roll off

CMRR 50Hz typically -93dB @ 20dB gain, -98dB @ 62dB gain;

20kHz signal typically -61dB @ 20dB gain, -100dB @

62dB gain

'Some of the factors which make the DACS MicAmp2 sound so good ("...great detail, depth and clarity in the sound with a wonderful easy quality to it" as Calum Malcolm told us) are not easily translatable into performance figures. The three ultra low noise super matched pair transistors on the input stage have a lot to do with it. The microphone is connected directly to the input transistor array, giving the correct load without the addition of other components, which degrade the signal. They are current devices, so there are always lots of electrons moving around. We believe that this is the reason why, even at very low levels, there is never any graininess in the sound, and that it is always clear, open and detailed. The noise performance may most usefully be expressed in comparison with an NE5534, the low noise IC used in most professional applications. The MicAmp2's noise is typically 16dB less, even at maximum gain.

# Compliance to European Standards

This MicAmp2 complies with the following standards (see Declaration of Conformity):

Radiated Emissions to Specification EN50081-1 Conducted Emissions to Specification EN50081-1 Electro Static Discharge to Specification EN50082-1 Fast Burst Transients to Specification EN50082-1

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