

# The Dangerous 2-BUS *LT* Manual

Thank you for choosing products from the exciting line of *Dangerous* recording equipment. Many years of dependable and trouble-free performance can be expected from our gear. This has been made possible by the careful design, construction and top-shelf component choices by recording industry veterans. The designers here at *Dangerous* are committed to a common goal: to bring you the highest quality possible for your dollars.

This manual will assist you in the installation of the 2-BUS *LT*.

## Table of Contents

<b>Overview .....</b>	<b>3</b>
<b>Hook-up .....</b>	<b>3</b>
<b>Usage Examples.....</b>	<b>4</b>
<b>Studio Calibration.....</b>	<b>5</b>
<b>Internal Jumpers.....</b>	<b>6</b>
<b>Grounding Examples .....</b>	<b>7</b>
<b>Voltage, dB, Bits.....</b>	<b>11</b>
<b>2-BUS <i>LT</i> Specifications.....</b>	<b>12</b>
<b>Connector Pinouts.....</b>	<b>14</b>

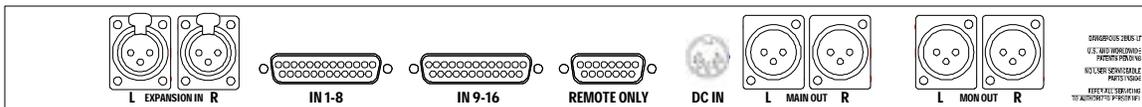
Certain precautions should be taken when using electrical products. Please observe the safety hints by reading the manual, following the directions, and obtaining qualified help if necessary to adhere to the precautions.

1. Use a properly grounded power supply with this product. Please do not defeat the ground pin on the mains plug. This connection protects the user in case of an internal fault by providing a low impedance path to ground. If a ground lift mains adapter is necessary to avoid hum problems, the system setup should be reviewed to locate the ground loop. The chapter on “Grounding examples” gives an explanation on obtaining quiet, safe system performance.
2. Avoid high temperature operation by providing air circulation space in equipment racks. The number one killer of electronic gear is HEAT. Placing 2-BUS *LT* in a hot rack with no ventilation will reduce the life span significantly depending on the operating temperature. This rule applies to the other electronic equipment in your racks also.
3. Avoid areas of high magnetic fields. The steel chassis of 2-BUS *LT* is designed to shield the circuits from EMI and RFI (magnetic and radio interference). When installing equipment in racks, it is advisable to put power amplifiers and other high powered equipment at least several rack spaces away from microphone pre-amps and summing amplifiers. Separation of high level and low level equipment can preempt trouble caused by heat and EMI.
4. Care should be taken to avoid spills of liquids around electrical gear. If a spill occurs, please shut off the equipment. A qualified technician should investigate accidents to prevent further equipment damage or personnel hazards caused by spills.
5. Every attempt has been made by the designers to provide versatility in the setup and calibration of this product. As such, there ARE user serviceable parts inside (what a pleasant surprise). If one is uncomfortable with opening up gear and changing jumpers or making adjustments, please seek qualified help if necessary.

## Overview

The 2-BUS *LT* is a 16X2 summing amplifier designed to help the users of digital audio workstations achieve better mix performance through the use of the existing equipment in their studios. The designers and their colleagues have noted that while digital audio workstations (DAW's) offer unprecedented flexibility in multitrack recording and editing, the mixing buss in these systems generally doesn't perform up to the quality of high end analog recording consoles in terms of sound quality and preservation of spatial detail. In today's portable environment, and with the cost of maintaining and housing legendary recording equipment, the choice of the big mixing desk is impossible for many users of DAW's. It is in this spirit that the Dangerous engineering team is designing and manufacturing an exciting array of equipment designed to meet the challenges of today's recording environment.

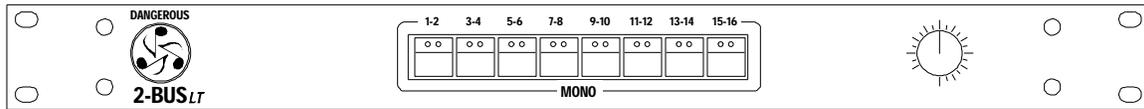
## Hooking up your Dangerous 2-BUS *LT*



DANGEROUS 2-BUS LT REAR PANEL

The 2-BUS *LT* is designed to mix the outputs of one or two eight channel D/A converters (Digidesign 882, 888, 192HD, Apogee AD8000, Prism Dream ADA-8, RME ADI series, Steinberg Nuendo 8, Troisi DC8-224DAC, etc.) The hook-up is simple. The *LT* employs what is known as the “DA-88” standard pinout for its analog inputs (see pg. 12.) Use a cable with the appropriate connectors to plug from the D/A analog output to the 2-BUS *LT* inputs. We recommend Redco Audio for cables and harnesses. We have specified a couple for our gear, to make life easier for those of you in need of harnesses (see also pg. 12.) The expansion inputs are for stacking 2-BUS's, or as an effects return to the mix. The “Main” output XLR's feed your 2-track mixdown medium, be it an external A/D converter, DAT machine, CD writer, analog 2-track, etc. The “Monitor” outputs go to an analog input on a *Dangerous* Monitor box or to the monitor section of an existing console. The levels of both 2-BUS *LT* outputs are at a nominal +4 dBu studio line-level (1.23 volts RMS). It is sometimes desirable to lift the shield on the receiving end of an output cable to ensure the best noise performance. Using “ground lift” plugs on equipment is generally not the best way to achieve quiet performance. The chapter titled “Grounding Examples” goes over some of the principles of this subject and is recommended reading for anyone setting up a studio. The Grounding Examples chapter also has wiring diagrams to assist users with unbalanced gear should they encounter trouble.

## Usage Examples



DANGEROUS 2-BUS LT FRONT PANEL

There are two types of controls on the 2-BUS *LT*- namely the “Mono” switches, and a potentiometer to control the output gain. The switches on the front panel on a 2-BUS *LT* enable the inputs to be treated as stereo pairs (panned hard left-right) or as individual inputs panned up the middle. This function is selected by pushing the “Mono” button of any channel pair to put both inputs of the pair in the center of the mix. For sounds that are panned somewhere in between the two extremes, a pair of outputs is used, the panning done in the computer, and the “Mono” button is not engaged.

For example, let’s say that one wants to mix a drum kit with bass, guitars, a lead vocal, and reverb on all of the above. To keep things simple, assume that our engineer has a Pro-Tools system with one 888 I/O box for a total of 8 analog feeds into the 2-BUS *LT*. This typical mix has 8 tracks of drums, one bass, two guitars, one vocal, and a reverb return for a total of fourteen tracks in the system. Usually, a DAW user would mix the song then spit the resulting stereo feed to two additional tracks internally or out of the digital output of the 888 onto a DAT machine. By using 4 sets of outputs in Pro-Tools, the audio performance can be increased by assigning the drums to PT outputs 1&2, the bass and vocal to outputs 3&4, and the guitars to outputs 5&6. The bass and vocal want their own channels so they are assigned individual outputs in Pro-Tools and the “Mono” button for channels 3-4 is pressed on the front panel of the 2-BUS *LT* to put the two “up the middle.” Reverb is sent from the tracks to a plug-in and the result routed to outputs 7&8. The panning on the drums and guitars is determined by the setting of the panners on the Pro-Tools desk. The “Mono” button pans the bass and vocal to the center of the mix. The “Main” output is sent to an external A/D converter then to a DAT machine for recording. This mix has the same balance as the one done internally, but because the “stems” can be run hotter onto separate busses, and summing to a stereo pair is being done externally in a high-quality analog environment, more of the original track’s detail winds up in the mix. The results are excellent imaging, clarity, punch, and better overall musicality. Breaking out into multiple analog outputs also permits the use of external analog compressors, EQ’s, and other effects

processing. Being able to process individual tracks, stems, and the entire mix with hardware means that you don't have to deal with the latency delay problems of DSP. You can concentrate on the music.

The output "Gain" control has a range of roughly 10 dB to allow hitting a following A/D or 2-track just right. Full clockwise is unity gain, but 2-BUS *LT* likes to run around 12:00 in a typical mix. This allows one to run the stems hotter than allowed in the DAW's digital 2-mix buss, resulting in less bit loss and increased preservation of the track's original sound.

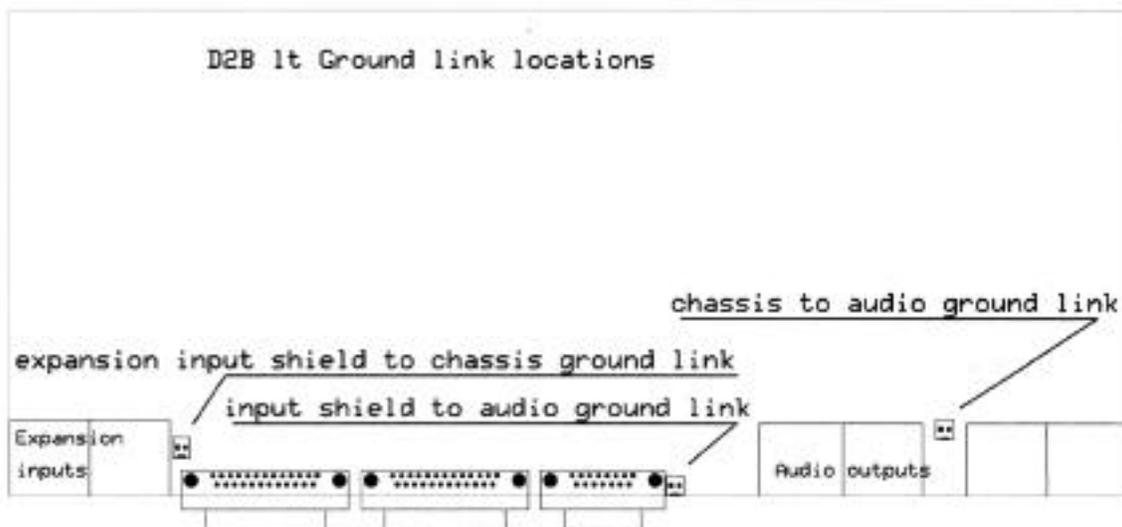
*The following chapters provide a more detailed look at some aspects of recording studio setup and wiring.*

## **Studio Calibration**

In order to enjoy the full benefits of the 2-BUS *LT*, it is necessary to align the D/A converters that feed it. This can be done with the aid of an AC voltmeter (available from Digi-Key, Techni-Tool, Radio Shack, et.al.) and the digital oscillator found inside the DAW. A known accurate VU meter, if available (patchable) is even easier to use. A 1 kHz signal at a level of -14dB full scale should be fed to the individual channels of the D/A converters from the DAW oscillator and the voltage at the output connectors adjusted to read 1.23 volts AC (+4dBu) across pins 2&3 of the XLR on professional level systems. Semi-pro systems (ones with RCA connectors,) if adjustable, usually are set to .245 volts RMS (-10dBu) measured between the center pin and the shell of the connector. (If there are no adjustments for level on the device then it is best not to worry as long as the levels are the same within 1/4 dB or several millivolts.) This adjustment sets the maximum operating level from the converters to +18dBu (on balanced D/A's) and will avoid clipping professional outboard gear. The 2-BUS *LT* inputs clip above +24dBu so headroom is not a problem on individual inputs.

## Internal Jumpers

2-BUS *LT* is shipped from the factory with its internal jumpers set to interface with balanced equipment. There are rare cases that certain pieces of gear might buzz with the factory presets. If this happens, the following diagrams and chapter have information that can help the user achieve quiet performance.



Ground header jumpers are included on the main circuit board to facilitate proper system grounding with 2-BUS *LT*. Please unplug the 2-BUS *LT* if you elect to open the case. Dropped screws on a live circuit board are not a pretty sight and the result is easy for the service department to figure out (non-warranty damage.) Depending on the internal wiring, a buzz can occur when using unbalanced D/A converters and may be cleared up by moving the header jumper marked “input shield to audio ground link” on the drawing so that the jump is connected. A buzz that may occur with an unbalanced feed into the “expansion input” XLRs may be cleared up by linking that header. The “chassis to audio ground link” will separate the box from the board without resorting to a ground lift AC plug. If moving the jumpers doesn’t clear up a buzz, disconnect any installed cables one at a time to determine where the problem is. The following chapter discusses system grounding in more detail to help troubleshoot interesting studio problems.

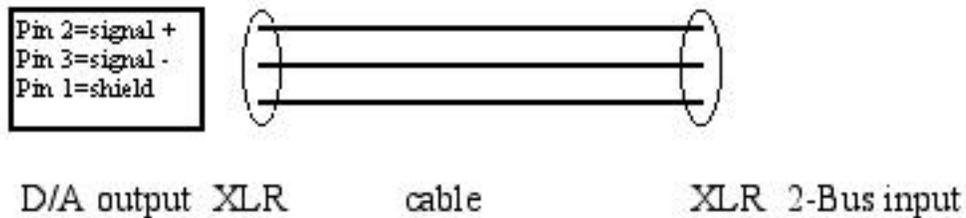
## Grounding Examples

To achieve maximum performance from the 2-BUS *LT*, cables need to be wired correctly. Store-bought cables can work just fine. Custom cable sets can be ordered from Redco, Inc. (see the last chapter for part numbers and ordering information.) Below are diagrams of frequently encountered wiring scenarios to help explain some of the possibilities. We'll start with professional, balanced, +4dBu systems encountered in most studio situations and then cover some contingencies that happen when interfacing -10dBu, semi-pro, unbalanced equipment.

The balanced world (XLR- type connectors) is the easiest to deal with because the audio signals and shield are treated separately down the cable.

Properly wired systems that use balanced interfaces can achieve impressive signal to noise ratios with the least amount of bother. Unbalanced equipment (RCA and two-wire 1/4" type connectors) can be made to perform very well but sometimes requires more effort because the signal reference (low side, or ground) and the cable shield (and usually, the chassis) of such equipment is shared. This sharing can cause hum or buzzing problems for several different reasons that can usually be remedied with some logical thinking and judicious jumper placement, or soldering.

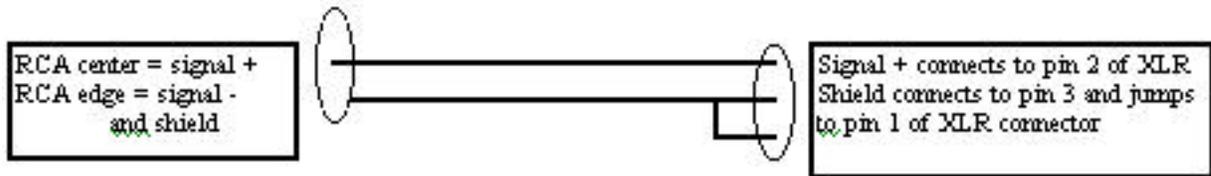
The 2-BUS *LT* inputs are designed to make the unbalanced interface easy to pull off properly if some simple rules are followed, but first, let's explore the balanced input interface.



Appropriate cables are used to plug the D/A into the 2-BUS *LT* input. Our diagrams show the use of an XLR breakout cable to illustrate the ideas. Shields should not be lifted as this is done at the appropriate place inside the 2-BUS *LT* to keep the input cable shield from conducting ground current. This is the beauty of the balanced connection. The signal pins carry the signal across them (transverse mode) and noise that gets through the shield is picked up equally by both signal conductors (if they are a twisted pair). This "common mode" noise is canceled by the differential action (subtraction) of the instrumentation amplifier in the first stage of the 2-BUS *LT*. Since the shield is not connected at both ends, current does not flow

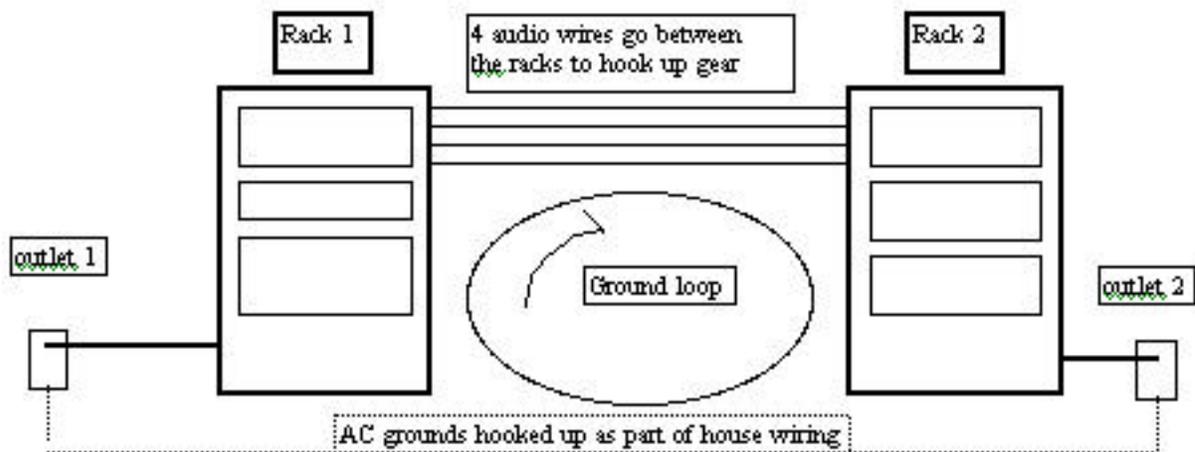
down the shield wire and no ground loop results from this interface. Audio goes through, noise is canceled, and grounds stay inside their respective pieces of gear. Beauty exists.

An unbalanced D/A driving a 2-BUS *LT* presents no problem due to the differential action of the input stage.



These two scenarios cover the receiving side of the 2-BUS *LT*. The driving side uses the same principles but optimum system performance usually requires the use of the dreaded voltmeter plus possibly a soldering gun. Don't worry, read on. Let's look at the scenario where a 2-BUS *LT* drives the balanced input of an A/D converter (either on the DAT machine, or a separate component).

Equipment manufacturers are required by CE standards to connect pin 1 of an input XLR to chassis ground at the connector in order to get the coveted sticker that lets them sell gear in Europe. This can cause ground loops if the shield wire is allowed to hook both ends of the cable up. The problem stems from the fact that two grounds in a system are never at the same potential. They can be close if the two pieces of gear in question are in the same rack or a heavy gage wire is used to bolt both the chassis together. Some people in desperation resort to using AC plug "ground lifts" to defeat the safety grounds (the third pin on an AC cord) in a random fashion until the system quiets down. This in our view is an unacceptable method of curing ground loop buzzes or hums. The diagram illustrates the problem and a solution follows.



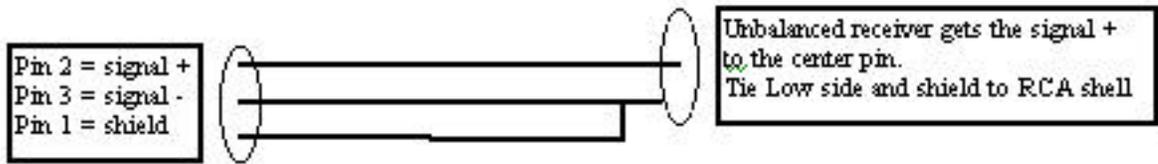
If the audio cables between the racks connect the grounds together via the shield pins, and the racks are at even slightly different potentials (on different power circuits, one draws heavy juice with amplifiers, long distance from each other, etc.) the shield of the audio cable will try to equalize the potential difference between the two racks. Juice will flow down the shield of the audio cable and broadcast hum into the signal wires the shield was supposed to protect. This situation manifests itself as the all too familiar buzz of a ground loop. The intensity depends on many variables but can go from unnoticeable to raging. The simple way to avoid this problem is take the voltmeter and switch it to measure continuity (the unit beeps when the leads are touched together).

Pull the AC plug, disconnect the cables from the back of the A/D and put one of the leads into pin 1 of the input XLR connector. Touch the other lead to the AC ground pin or a convenient chassis screw and see if the VOM beeps. If it does, then pin 1 is grounded and the shield of the interconnecting cable (pin 1) should be unsoldered and taped (so as not to inadvertently short to the shell of the XLR) at the male connector.

Some equipment manufacturers provide a header inside the gear to lift the pin 1 connection. In this case the VOM in the above test would not beep and the audio cable can be left alone. This test can be performed on most balanced line level interfaces to clear up ground loops without applying AC ground lift adapters (which are generally unsafe to use unless the gear in question is grounded by some other method). If you are unfamiliar with VOM and soldering techniques, this whole procedure can be avoided and the system just hooked up to hope for the best. In small installations where there are no great distances for signals to travel or everything is in the same rack, resulting ground currents cause hum below the desired noise floor.

Indications on the A/D meters of greater than -70 dB (-60dB? you be the judge) with no music going should be cause to pick up the phone and get some help from a friend who knows how to use a VOM. It's really not that hard and the extra work can make a big difference in the quality of your mixes. The self noise of a Dangerous 2-BUS *LT* is better than -80dB in a bandwidth of 22Hz-22 kHz, and with the A/D aligned at -14dBfs for an input of +4dBu provides a better than 98dB dynamic range for clean, quiet, punchy recording, yeah baby.

Driving an unbalanced input of an A/D is done by using balanced cable at the 2-BUS *LT* end and tying pins 1&3 together at the A/D end and applying that connection to the shell of the RCA connector as in the following diagram.



If noise is encountered using an unbalanced connection, the following steps will clear up the problem. It may be necessary to bolt the A/D in the same rack as the 2-BUS *LT* to obtain optimum performance. Not being able to run long lines is the potential downside of unbalanced equipment, but this method can be effective. Isolating the A/D from the rack would be the next step. Using an AC ground lift adapter and letting the offending A/D (or DAT machine) get its ground down the signal line would be the last resort although this technique can still work OK. Some DAT decks with unbalanced inputs have two-pronged power cords and electrically isolating them from the rack and providing ground down the audio cable is a situation that has worked well and is really functionally equivalent to the last resort mentioned above. Use the VOM set to continuity mode to check isolation from the rack (with all the cables unplugged from the deck while testing).

## Volts, Decibels, Bits (16), and Bits (24)

	Volts	dBu	(16)Bits	(24)
The voltage list is for a sine wave AC	6.20	+18	16	24
measured root mean squared (the way a	3.10	+12	15	23
volt meter would). The numbers are	1.55	+6	14	22
rounded off so a direct comparison to a	1.228	+4	<b>13+</b>	<b>21+</b>
calculation could be slightly different.	0.775	0dBu	13	21
Professional levels in a studio are	0.3875	-6	12	20
referenced to 1.228VACRMS. This is	0.245	<b>-10</b>	<b>11+</b>	<b>19+</b>
the voltage out of a console when a VU	0.1938	-12	11	19
meter says "0dB". The dB scale is a	0.0969	-18	10	18
logarithmic scale that is easier to deal with	0.04844	-24	9	17
in a working world of audio than a voltage	0.02422	-30	8	16
scale because loudness in hearing also	0.01211	-36	7	15
follows a log scale. 6dB is twice the	0.00605	-42	6	14
voltage. 10dB is roughly twice the loudness.	0.00303	-48	5	13
20dB is 10 times the voltage. The 16 bits	0.00151	-54	4	12
column represents the number of bits on a	0.000757	-60	3	11
CD or the number of bits that it takes to	0.000378	-66	2	10
represent the voltage at a given level.	0.000189	-72	1	9
The 24 bit column represents how a	0.000095	-78	dither	8
theoretical 24 bit converter would	0.000047	-84		7
represent the given voltage. I don't	0.000024	-90		6
know of any microphone preamp,	0.000012	-96		5
console, or A/D converter that can deliver	0.000006	-102		4
144dB of dynamic range but perhaps it	0.000003	-108		3
is good to represent the noise floor	0.0000014	-114		2
with several bits. It is certainly not a bad	0.0000007	-120		1
idea to process digital audio information		-126	dither	
with at least a 24 bit system if number				
manipulation (DSP) is taking place.				

In a digital system that uses binary counting (PCM being popular) every bit doubles the number of the possible voltages that can be present at the output of a D/A converter. A 16 bit number represents 65536 different possible voltages in a PCM converter assuming no DC offset and a noise floor at the least significant bit. The smallest step is roughly 94uV. Stack 65536 90uV steps on top of one another and you get the 6.2 volt maximum. The step size for a 24 bit word under these same conditions is about .7uV, a small step indeed. The point of this exercise is to illuminate the relationship between voltage, dB's, and bits in a PCM system.

## **2-BUS *LT* Specifications**

*Measurements taken at +4 dBu nominal level*

Frequency Response	1 Hz- 100 kHz within 0.3dB
Total Harmonic Distortion	0.006% in audio band
Intermodulation Distortion	0.007% IMD60 4:1
Crosstalk @ 1 kHz	-92 dB
Noise Floor	-81 dBu total energy in audio band
Maximum level	+24 dBu
Nominal operating level	+ 4 dBu
Input impedance	25k ohms balanced
Output impedance	50 ohms (600 ohm drive capable)
Power consumption	25 watts
Warranty	2 years parts and labor. Subject to Inspection. Does not include shipping Damage, abusive operation, or modifications/attempted repair by unauthorized personnel.

Please fill out the registration card within two weeks of purchase to help us keep our paperwork straight. Any information included on the card is treated as personal and will not be used in any mailings or disclosed to third parties.

The warranty is activated if the card is filled out and mailed to:

Dangerous Music, Inc.  
154 east 2<sup>nd</sup> street #4  
New York, NY 10009  
[www.dangerousmusic.com](http://www.dangerousmusic.com)

In case of suspected trouble with a DM Inc. unit, check the operation manual to see if the problem is addressed by internal jumper settings or usage suggestions. If not, please write or email the factory at the above address for service information. It is recommended to save the original shipping box for moving, service or storage. The warranty applies to the original purchaser.

Model # \_\_\_\_\_

Serial # \_\_\_\_\_

Inspected \_\_\_\_\_

## Connector Pin outs

Power connector: 5 pin DIN

Pin1 ground  
Pin 2 ground  
Pin 3 +5V DC  
Pin 4 -15V DC  
Pin 5 +15V DC

XLR connectors

pin 1 shield  
pin 2 + audio  
pin 3 - audio

25 pin D connectors (audio inputs)

<u>pin#</u>	<u>channel#</u>	<u>polarity</u>
1	8	+
14	8	-
2	8	shield
15	7	+
3	7	-
16	7	shield
4	6	+
17	6	-
5	6	shield
18	5	+
6	5	-
19	5	shield
7	4	+
20	4	-
8	4	shield
21	3	+
9	3	-
22	3	shield
10	2	+
23	2	-
11	2	shield
24	1	+
12	1	-
25	1	shield
13	no connection	

This is the same pinout used by DA-88 decks and many other audio devices. Several lengths are available – give Redco a call or visit the website (800-572-7280, [www.redco.com](http://www.redco.com))

XLR to 25-pin D	part# DM-1
25-pin D to 25-pin D	part# DM-2
RCA to 25-pin D	part# DM-3