



I-Tech 4x3500HD DriveCore™ Series

Operation Manual



Product Registration: Register your new product at <http://warranty.harmanpro.com>.

Obtaining Other Language Versions: To obtain information in another language about the use of this product, please contact your local Crown Distributor. If you need assistance locating your local distributor, please contact Crown at 574-294-8000 or www.crownaudio.com.

This manual does not include all of the details of design, production, or variations of the equipment. Nor does it cover every possible situation which may arise during installation, operation or maintenance.

The information provided in this manual was deemed accurate as of the publication date. However, updates to this information may have occurred. To obtain the latest version of this manual, please visit the Crown website at www.crownaudio.com.

Trademark Notice: Crown, Crown Audio, IQ, BCA, and Amcron are registered trademarks of Crown International. HiQnet is a trademark of Harman International Industries, Inc. Other trademarks are the property of their respective owners.

Some models may be exported under the name Amcron.®

Important Safety Instructions

Importantes Instructions de Sécurité

1. Read these instructions.
2. Keep these instructions.
3. Heed all warnings.
4. Follow all instructions.
5. Do not use this apparatus near water.
6. Clean only with a dry cloth.
7. Do not block any ventilation openings. Install in accordance with the manufacturer's instructions.
8. Do not install near any heat sources such as radiators, heat registers, stoves, or other apparatus (including amplifiers) that produce heat.
9. 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 grounding-type plug has two blades and a third grounding prong. The wide blade or the third prong is provided for your safety. If the provided plug does not fit into your outlet, consult an electrician for replacement of the obsolete outlet.
10. Protect the power cord from being walked on or pinched, particularly at plugs, convenience receptacles, and the point where they exit from the apparatus.
11. Only use attachments/accessories specified by the manufacturer.
12. Use only with a cart, stand, tripod, bracket, or table specified by the manufacturer, or sold with the apparatus. When a cart is used, use caution when moving the cart/apparatus combination to avoid injury from tip-over.
13. Unplug this apparatus during lightning storms or when unused for long periods of time.
14. Refer all servicing to qualified service personnel. Servicing is required when the apparatus has been damaged in any way, such as power-supply 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.
15. Use the mains plug to disconnect the apparatus from the mains.
16. WARNING: TO REDUCE THE RISK OF FIRE OR ELECTRIC SHOCK, DO NOT EXPOSE THIS APPARATUS TO RAIN OR MOISTURE.
17. DO NOT EXPOSE THIS EQUIPMENT TO DRIPPING OR SPLASHING AND ENSURE THAT NO OBJECTS FILLED WITH LIQUIDS, SUCH AS VASES, ARE PLACED ON THE EQUIPMENT.
18. THE MAINS PLUG OF THE POWER SUPPLY CORD SHALL REMAIN READILY OPERABLE.



Wichtige Sicherheitsinstruktionen

Instrucciones de Seguridad Importantes

- TO PREVENT ELECTRIC SHOCK DO NOT REMOVE TOP OR BOTTOM COVERS. NO USER SERVICEABLE PARTS INSIDE. REFER SERVICING TO QUALIFIED SERVICE PERSONNEL.
- À PRÉVENIR LE CHOC ÉLECTRIQUE N'ENLEVEZ PAS LES COUVERCLES. IL N'Y A PAS DES PARTIES SERVICEABLE À L'INTÉRIEUR. TOUS REPARATIONS DOIT ÊTRE FAIRE PAR PERSONNEL QUALIFIÉ SEULEMENT.
- PARA PREVENIR UN CHOQUE ELÉCTRICO, NO RETIRE LAS CUBIERTAS SUPERIOR O INFERIOR. NO EXISTEN PARTES QUE PUEDAN SER REPARADAS POR EL USUARIO AL INTERIOR. REMITA EL SERVICIO AL PERSONAL TÉCNICAL CALIFICADO.
- TO COMPLETELY DISCONNECT THIS EQUIPMENT FROM THE AC MAINS, DISCONNECT THE POWER SUPPLY CORD PLUG FROM THE AC RECEPTACLE. THE MAINS PLUG OF THE POWER SUPPLY CORD SHALL REMAIN READILY OPERABLE.
- POUR DÉMONTER COMPLÈTEMENT L'ÉQUIPEMENT DE L'ALIMENTATION GÉNÉRALE, DÉMONTER LE CÂBLE D'ALIMENTATION DE SON RÉCEPTEACLE. LA PRISE D'ALIMENTATION RESTERA AISÉMENT FONCTIONNELLE.
- PARA DESCONECTAR COMPLETAMENTE EL EQUIPO DEL SUMINSTRO ELECTRICO, DESCONECTE EL CABLE DE ALIMENTACION DE LA TOMA DE CA. LAS PATAS DEL CONECTOR DEL CABLE DE ALIMENTACIÓN DEBERAN MANTENERSE EN BUEN ESTADO.
- WATCH FOR THESE SYMBOLS:**
- The lightning bolt triangle is used to alert the user to the risk of electric shock.
- The exclamation point triangle is used to alert the user to important operating or maintenance instructions.
- REGARDEZ CES SYMBOLES:**
- La triangle avec le sigle "foudre" est employée pour alerter l'utilisateur au risque de décharge électrique. Le triangle avec un point d'exclamation est employée pour alerter l'utilisateur d'instruction importantes pour lors opérations de maintenance.
- ATENCIÓN CON ESTOS SÍMBOLOS:**
- El triángulo con el símbolo de rayo eléctrico es usado para alertar al usuario de el riesgo de un choque eléctrico.
- El triángulo con el signo de admiración es usado para alertar al usuario de instrucciones importantes de operación o mantenimiento.



IMPORTANT

I-Tech Series amplifiers require Class 2 output wiring.
Les amplificateurs de série de I-Tech exigent des câbles de sortie de classe 2.

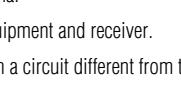
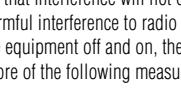
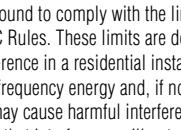
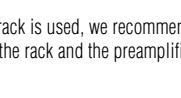
I-Tech-Reihe-Verstärker verlangen Klasse die 2 Produktionsverdahrtung.

Los amplificadores de la Serie I-Tech requieren de un cableado de salida Clase 2.

MAGNETIC FIELD

CAUTION! Do not locate sensitive high-gain equipment such as pre-amplifiers directly above or below the unit. Because this amplifier has a high power density, it has a strong magnetic field which can induce hum into unshielded devices that are located nearby. The field is strongest just above and below the unit.

If an equipment rack is used, we recommend locating the amplifiers in the bottom of the rack and the preamplifier or other sensitive equipment at the top.



The I-Tech Series amplifiers are certified only at 120V ~ in Canada.

CAUTION

AVIS

MAGNETIC FIELD

NOTE:

CAUTION:

NOTE:

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1 Welcome

The Crown® I-Tech HD DriveCore Series offers amazing power, light weight and ease of use for touring sound applications. Unlike other amplifiers, it includes onboard high-definition DSP, a 4.3" LCD screen, and a built-in network connection.

Modern power amplifiers are sophisticated pieces of engineering capable of producing extremely high power levels. They must be treated with respect and correctly installed if they are to provide the many years of reliable service for which they were designed.

In addition, I-Tech Series amplifiers include a number of features which require some explanation before they can be used to their maximum advantage.

Please take the time to study this manual so that you can obtain the best possible service from your amplifier.

1.1 Features

- Global Power Supply with PFC (Power Factor Correction) works anywhere in the world.
- High power density, up to 12,000 watts in a 2U chassis.
- Output voltage of 185Vpk provides clean transient peaks.
- 6th-generation patented Class I (BCA® with Drivecore Technology) circuitry couples power efficiently to the load and provides low AC current draw.
- Front-panel USB connector accepts a USB drive to transfer presets from the drive to the amplifier DSP, and vice versa. This feature also allows you to update the amplifier's firmware.
- Light weight due to aluminum chassis, special internal construction, switching power

supply and patented class-I output stage with DriveCore Technology.

- Thermal management controller and two discrete thermal zones with variable-speed fans, forced-air cooling.

• Advanced protection circuitry guards against: shorted outputs, DC, mismatched loads, general overheating, under/over voltage, high-frequency overloads and internal faults.

• Five-Year, No-Fault, Fully Transferable Warranty completely protects your investment and guarantees its specifications.

1.2 How to Use This Manual

This manual provides you with the necessary information to safely and correctly setup and operate your amplifier. It does not cover every aspect of installation, setup or operation that might occur under every condition. For additional information, please consult the online help in System Architect software, Crown's *Amplifier Application Guide*, *I-Tech Application Guide* (available online at www.crownaudio.com), Crown Technical Support, your system installer or retailer.

We strongly recommend you read all instructions, warnings and cautions contained in this manual. Also, for your protection, please send in your warranty registration card today. And save your bill of sale — it's your official proof of purchase.

2 Setup



2.1 Unpack and Install Your Amplifier

Please unpack and inspect your amplifier for any damage that may have occurred during transit. If damage is found, notify the transportation company immediately. Only you can initiate a claim for shipping damage. Crown will be happy to help as needed. Save the shipping carton as evidence of damage for the shipper's inspection.

We also recommend that you save all packing materials so you will have them if you ever need to transport the unit. **Never ship the unit without the factory pack.**

YOU WILL NEED (not supplied):

- Input wiring cables
- Output wiring cables
- Ethernet cables
- Rack for mounting amplifier (or a stable surface for stacking)

WARNING: Before you start to set up your amplifier, make sure you read and observe the Important Safety Instructions found at the beginning of this manual.

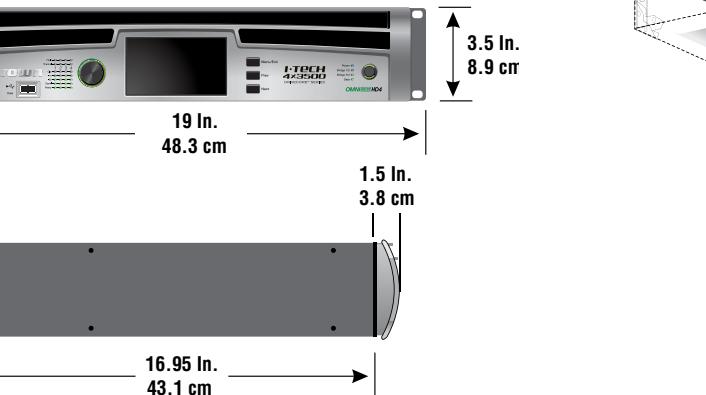


Figure 2.1
Dimensions



CAUTION: Before you begin, make sure your amplifier is disconnected from the power source, with the power switch in the "off" position and all level controls turned completely down (counterclockwise).

Use a standard 19-inch (48.3 cm) equipment rack (EIA RS-310B). See Figure 2.1 for amplifier dimensions.

You may also stack amps without using a rack.

NOTE: When transporting, amplifiers should be supported at both front and back.

MAGNETIC FIELD



CAUTION! Do not locate sensitive high-gain equipment such as preamplifiers or tape decks directly above or below the unit. Because this amplifier has a high power density, it has a strong magnetic field which can induce hum into unshielded devices that are located nearby. The field is strongest on the right side and right bottom of the amplifier (facing the amplifier).

If an equipment rack is used, we recommend locating sensitive equipment at least 20 cm (8 inches) away from the amplifier.



Figure 2.2 Airflow

When using an equipment rack, mount units directly on top of each other. Close any open spaces in rack with blank panels. DO NOT block front or rear air vents. The side walls of the rack should be a minimum of two inches (5.1 cm) away from the amplifier sides, and the back of the rack should be a minimum of four inches (10.2 cm) from the amplifier back panel.

Figure 2.2 illustrates standard amplifier airflow.

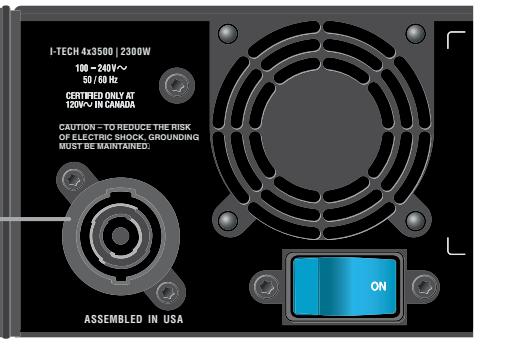
I-Tech HD DriveCore Series Power Amplifiers

2 Setup

2.2 Connecting to AC Mains

WARNING: The third (ground) prong of the supplied AC power cord connector is a required safety feature. Do not attempt to disable this ground connection by using an adapter or other methods.

Amplifiers do not create energy. The AC mains voltage and current must be sufficient to deliver the power you expect. You must operate your amplifier from an AC mains power source with not more than a 10% variation above or a 10% variation below the amplifier's specified line voltage range and within the specified frequency requirements (indicated on the amplifier's back panel label). If you are unsure of the output voltage of your AC mains, please consult your electrician.



32A Neutrik® powerCON

Figure 2.3 Power Connector

2.3 Wire Inputs and Outputs

2.3.1 Wiring basics

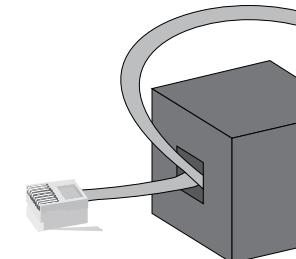
- Always use shielded wire for input wiring. The higher the density of the shield (the outer conductor) the better. Spiral wrapped shield is not recommended.
- When using unbalanced lines keep the cables as short as possible. Avoid lengths greater than 10 feet (3 meters).
- Do not run audio input cables together with high-level wiring such as loudspeaker wires or AC cords. (This lessens the chance of hum and noise being induced into the input cables.)
- Turn the entire sound system off before changing any connections. Crown is not liable for damage incurred when any transducer or component is over driven.

THE CHANNEL 2 AND 4 INPUT IS IGNORED by default in Bridge Mono mode. It can be summed using the input source selector.

For additional information on audio input wiring please refer to the Crown *Amplifier Application Guide* available online at www.crownaudio.com. It contains helpful information on preventing unwanted subsonic frequencies, radio frequency interference, ground loops, and feedback oscillation.

When using network connections, pass the CAT 5 cable five times through a ferrite core (Figure 2.4), available from Crown Audio Inc. This is to ensure compliance with emission regulations.

Figure 2.4 Pass the CAT 5 Cable Five Times Through the Ferrite Core



2 Setup

2.3.2 Choose Input Wire and Connectors

Crown recommends using pre-built or professionally wired, balanced line (two-conductor plus shield), 22-24 gauge cables and connectors. Use 3-pin male XLR connectors.

Unbalanced line may also be used but may result in noise over long cable runs.

Figure 2.5 shows connector pin assignments for balanced analog wiring or AES3 digital wiring. The use of standard analog cable with AES3 will result in diminished performance. For best results, 110 ohm shielded twisted-pair cable for AES3 signals is highly recommended. Figure 2.6 shows connector pin assignments for unbalanced analog wiring.

If multiple connectors are needed for AES3 distribution from a single AES3 source, the following shows the number of AES connections in parallel based on AES3 switching frequency.

Frequency	# of Connections
96kHz	4 AES3 Pairs
48kHz	8 AES3 Pairs
44.1kHz	8 AES3 Pairs

***Please note with good quality 110 ohm shielded digital cable, it is possible to obtain a higher number of connections than stated here. It is up to the user to test this based on their use case. The connections stated here are based on worst case analysis.

NOTE: Custom wiring should only be performed by qualified personnel.

2.3.3 Choose Output Wire and Connectors

Crown recommends using pre-built or professionally wired, high-quality, two-, four-, or eight-conductor, heavy gauge speaker wire and connectors. Use Class 2 output wiring. You may use a 4-pole Speakon® connector (Figure 2.7) or banana plugs, spade lugs, or bare wire for your output connectors (Figure 2.8). To prevent the possibility of short circuits, wrap or otherwise insulate exposed loudspeaker cable connectors.

CAUTION – SHOCK HAZARD: Potentially lethal voltages exist at the output connectors when the amplifier is turned on and is passing a signal.

Using the guidelines below, select the appropriate size of wire based on the distance from amplifier to speaker.

Distance	Wire Size
up to 25 ft.	16 AWG
26-40 ft.	14 AWG
41-60 ft.	12 AWG
61-100 ft.	10 AWG

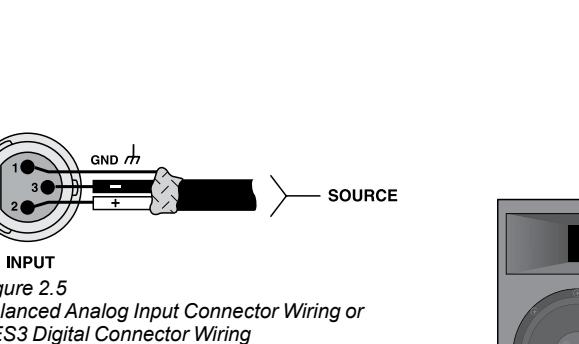


Figure 2.5
Balanced Analog Input Connector Wiring or
AES3 Digital Connector Wiring

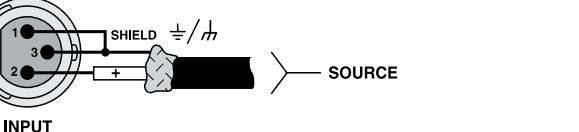


Figure 2.6
Unbalanced Analog Input Connector Wiring



Figure 2.7
Left: Speakon® Output Connector on Back Panel
Right: Speakon® Cable Connector

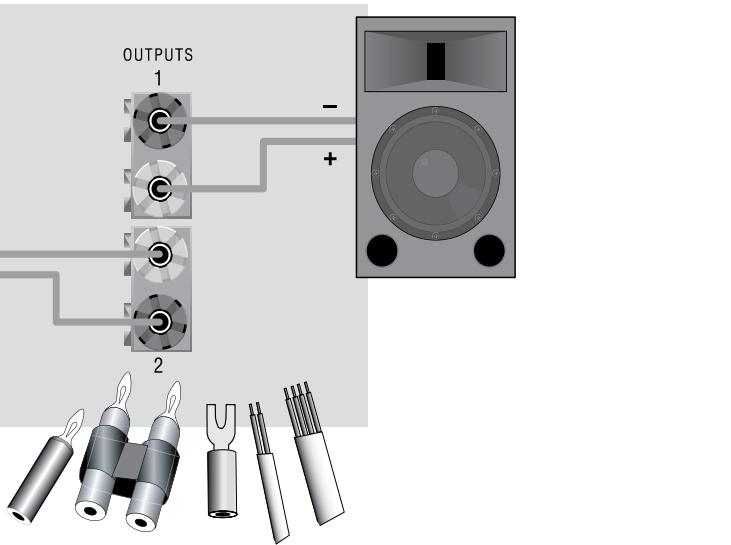


Figure 2.8
Binding Post Connections

2 Setup

101-150 ft. 8 AWG
151-250 ft. 6 AWG

CAUTION: Never use shielded cable for output wiring.

2.3.4 Stereo Mode Wiring

Typical input and output wiring is shown in Figure 2.9.

IMPORTANT: Turn off the amplifier and unplug its power cord.

INPUTS: Choose one of these options:

- Connect analog input wiring for all channels.
- Connect AES3 digital signal to the AES3 connectors.

OUTPUTS: Maintain proper polarity (+/-) on output connectors. Use Class 2 output wiring.

Figure 2.9 shows how to wire stereo speakers to the binding posts. Connect Channel 1 loudspeaker's positive (+) lead to Channel 1 positive (red) terminal of amp; repeat for negative (-). Repeat Channel 2 wiring as for Channel 1.

To wire stereo speakers to the Speakon® connectors, use this method:

Table 1
Stereo Wiring Method 1: Use Top Speakon® Only

PIN	1+	1-	2+	2-
CH	1+	1-	2+	2-

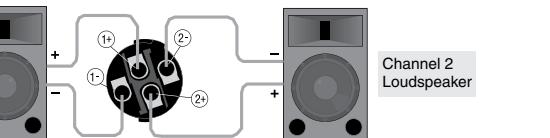


Figure 2.10
Wiring Two Stereo Speakers
to the Top Speakon® Connector

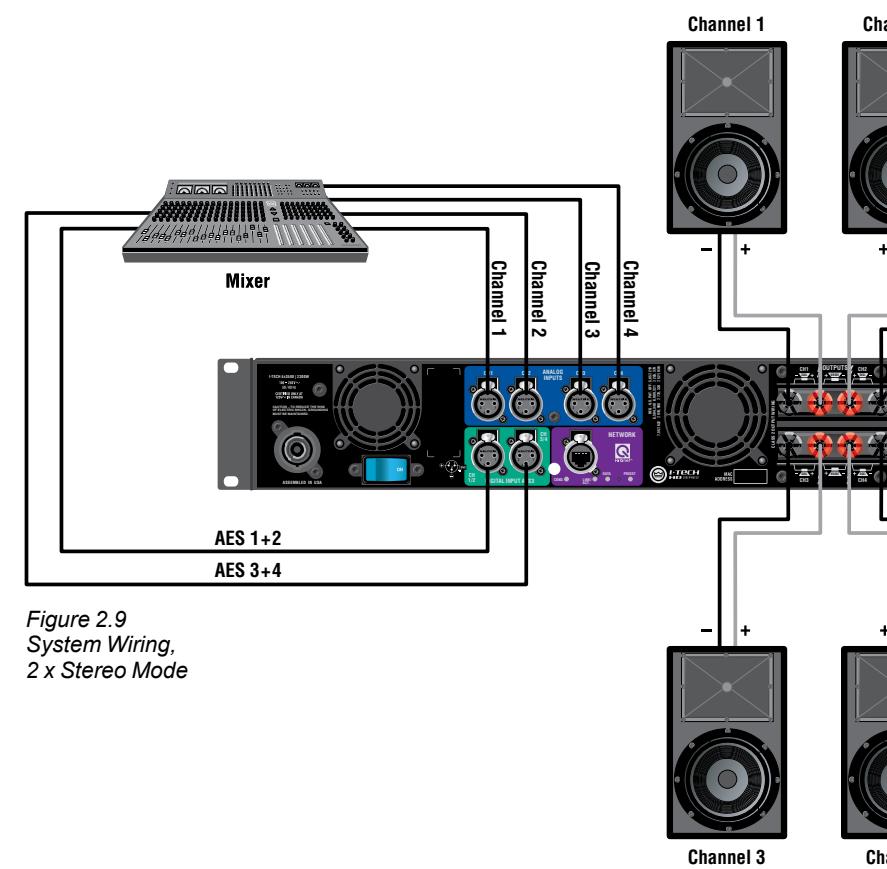


Figure 2.9
System Wiring,
2 x Stereo Mode

2 Setup

Table 1 and Figure 2.10: Wire one Speakon® cable connector to two speakers. Insert the Speakon® cable connector into the amplifier's top Speakon® connector.

2.3.5 Bridge-Mono Mode

Overview: Turn on the amp, enable Bridge-Mono mode using the LCD Control Screen, turn off the amp, wire it, and turn it back on.

1. Be sure that no cables are connected to the amplifier. Turn on the front-panel power switch. The LCD Control Screen will light up.

2. Under the LCD Control Screen, press the **Menu/Exit** button. Press the **Next** button until you see **OUTPUT MODE** on the screen. If **N/A** is displayed, **OUTPUT MODE** is locked via software. If **LOCKOUT** is displayed, all the LCD screens are locked via software.

3. Press the Encoder knob to select **BRIDGE MONO**. Press the knob again to confirm your choice. Press **Menu/Exit**. Turn down both level controls (Encoder) until you reach maximum attenuation.

4. IMPORTANT: Turn off the amplifier and unplug its power cord.

INPUTS: There are three ways to connect an input signal to the amplifier (Figure 2.11):

- Connect an analog signal source to the Channel-1 and/or Channel-3 amplifier input.

- Connect an AES/EBU digital signal source to the Digital Input IN connector.

NOTE: Crown provides a reference of wiring pin assignments for commonly used connector types in the *Crown Amplifier Application Guide* available at www.crownaudio.com.

OUTPUTS: Use Class 2 output wiring. There are two ways to wire the amplifier output connectors for Bridge-Mono mode:

1) Wire the speaker across the red binding post channels 1/2 and 3/4 (Figure 2.11). Do not use the black binding posts when operating in Bridge-Mono mode.

2) Wire the speaker only to the Speakon® connector as shown in Table 2 and Figure 2.12.

NOTE: In Bridge-Mono mode, the Channel 1 Level control sets the level; the Channel 2 Level control is defeated. All Channel 2 objects and controls are hidden and disabled. In this mode, the Channel 3 Level control sets the level; the Channel 4 Level control is defeated.

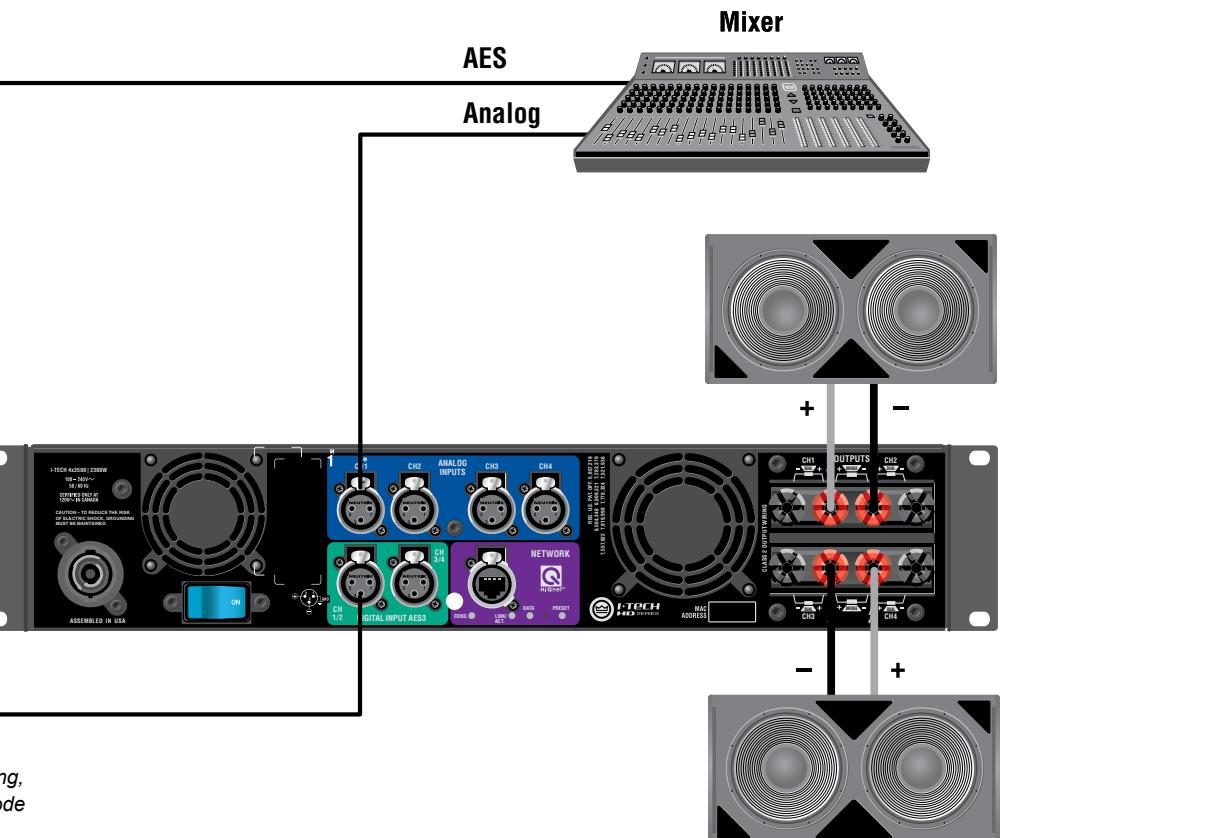


Figure 2.11
System Wiring,
2 x mono mode

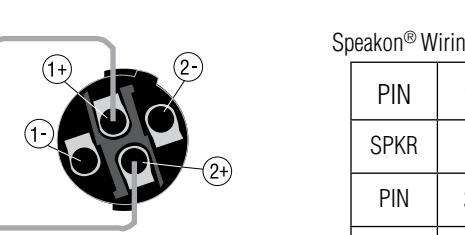


Figure 2.12
Mono Wiring Using Both

Table 2

Speakon® Wiring for Bridge-Mono

PIN	1+	2+
SPKR	+	-
PIN	3+	4+
SPKR	+	-

3 HARMAN GreenEdge -Going 'Green' without compromise

3.1 Going 'Green'

Going 'Green' is no longer an option – it's a necessity. Protecting the world we live in against adverse climate change is a major objective of socially responsible governments, businesses and individuals everywhere. Reducing energy consumption is fundamental to the reduction of greenhouse-gas emissions and is likely to impact various elements of our lifestyle. Maintaining the quality of our lifestyle, while improving the environment, is a challenge addressed by the HARMAN GreenEdge advanced technology initiative.

HARMAN's passion for innovative engineering remains dedicated to its goal of acoustic excellence for all of its products. HARMAN's loyal customers never need to fear the loss of performance from going 'Green' – in fact, the new range of GreenEdge audio systems enhance performance and acoustic qualities. HARMAN continues to drive the importance of 'the listening experience' while caring for the world we share.

3.2 The Sound of Efficiency

HARMAN's Research and Development teams are implementing highly innovative engineering, design and manufacturing processes across its range of world-class branded audio and infotainment products. The GreenEdge technology initiative allows partners to take advantage of Eco-Friendly, energy-saving benefits such as:

Weight Savings – When compared to traditional, non-switching amplifiers, the I-Tech HD Series weighs an average of 50 lbs (22.68kg) less. Less weight benefits the consumer by reducing shipping costs and benefits the environment by reducing emissions from vehicles because the shipper is able to either reduce the amount of vehicles needed to transport the goods or the vehicle is not running up to its weight limits resulting in less fuel economy.

High-Efficiency Designs – The patented Universal Power Supply with Power Factor Correction combined with the patented Class-I PWM output stage of the I-Tech HD series amplifiers provides greater than 90% efficiency when ran at 100VAC to 240VAC and all voltage in between.

Power Savings – The I-Tech HD Series has both manual and automatic standby modes that put the amplifier into a low AC current draw state when not in use. When in this mode, the I-Tech HD series draws an average of 100W of power greatly reducing your power bills.

Environmental Responsibility – By integrating HARMAN GreenEdge systems into your business model, you are promoting Eco-Friendly energy saving initiatives and reducing your impact on the environment.

3.3 Green Savings

Here's an example of the energy savings offered by Crown's Class-I tour products:

Class-I Amplifiers v. MA5002VZ in Touring Applications

- If Crown shipped 13,500 IT4000/MA5000/IT5000HD
- If we had shipped 13,500 MA5002VZs in their place, the numbers would look like this:

	MA5002VZ	Class-I Tour Amp (IT4000, MA5000, IT5000HD)
Number of Units	13,500	13,500
Total Weight	519.8 tons	189 tons
1 hour of Electricity Usage	2.52kWh	1,602kWh
1 year of Electricity Usage	18,370,800 kWh	11,678,580 kWh

Electricity Savings:

- ~6,692,220 kWh
- Enough to power ~370 households annually!

Weight Savings (Freight):

- Reduces ~400 Metric tons of greenhouse gas annually
- About the same as taking 75 cars off the road

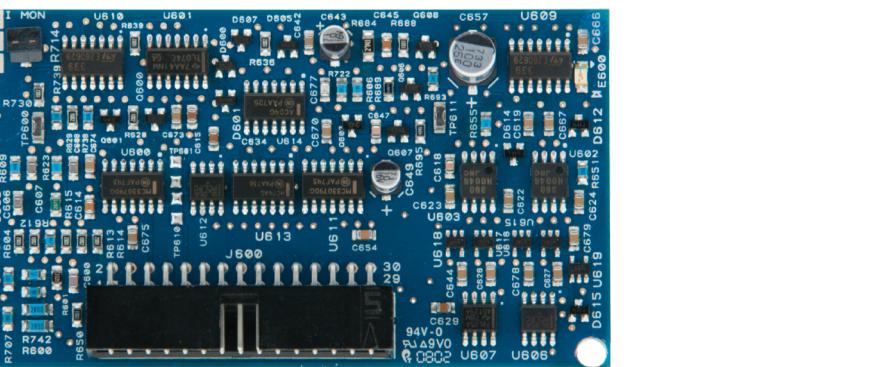
4 Integrated DriveCore™ Technology

Class D and Class I amplifiers are notable for extraordinarily high efficiency and being well suited for driving difficult reactive loads such as subwoofers.

Crown engineers developed DriveCore™ Technology – a proprietary hybrid analog-digital integrated circuit (IC) developed with Texas Instruments that drives the “front end” of the Class D or Class I output stage. Over 60 years of Crown’s design knowledge and experience went into the development of this technology resulting in truly remarkable benefits.

DriveCore™ Technology provides an extremely wide-tolerance with regards to sagging or "dirty" AC line conditions providing consistent performance without affecting audio quality. This means that your performance will not be compromised by fluctuating generator power or overloading by lighting rigs, backline gear, etc.

In addition, DriveCore™ Technology's patented feedback and PWM modulation circuits enable fast recovery on peak transients, accurate reproduction of low-level detail, and precise tracking of low-frequencies at high power levels for maximum subwoofer output.



1ch HD 2 Channel Front End Board

Copy Chip (Actual Size)

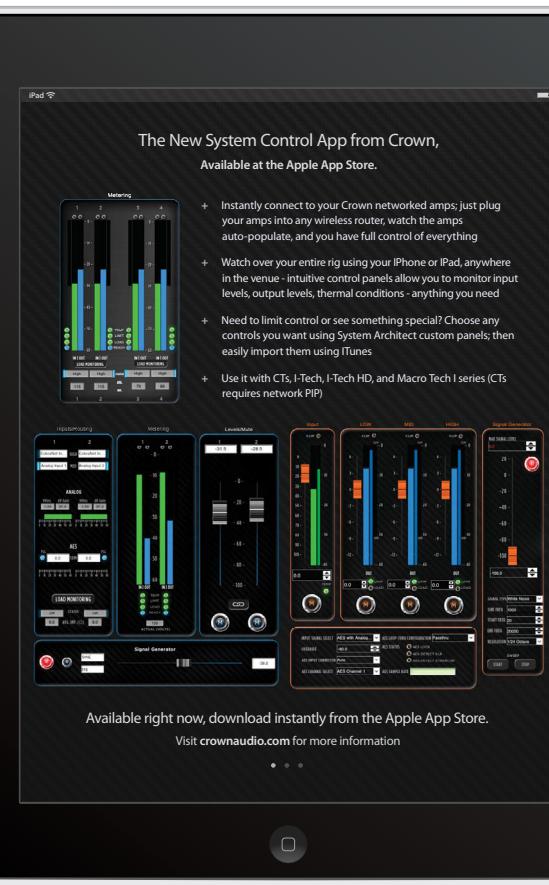
eCore Series Power Amplifiers

amplifier is compatible with the Powered by Crown
on visit:
www.crownaudio.com/mobileapplications/index.htm



My Crown

amplifier is compatible with the Powered by Crown
on visit:
www.crownaudio.com/mobileapplications/index.htm



6 Operation

6.1 Protecting Your Speakers

It's wise to avoid clipping amplifier signals at the input and output. Not only does clipping sound bad, it can damage loudspeakers. To prevent clipping, use System Architect software's Level Max suite to enable or display the peak voltage limiter and average power limiter in your amplifier's built-in DSP. That way, no matter how strong a signal your mixer produces, the amplifier output will not clip. Set the limiter threshold so that at full level no components in the audio signal chain are clipping, the amplifier input and output is not clipping, and the attached loudspeakers are properly protected according to the manfg's rec'd specifications.

Also, avoid sending strong subsonic signals to the amplifier. High-level, low-frequency signals from breath pops or dropped microphones can blow out drivers. To prevent subsonic signals, use one of these methods:

- Insert a highpass filter between mixer output and amplifier input (or between mixer and limiter).
- Use the I-Tech's onboard DSP to set up a highpass filters.
- Switch in highpass filters at your mixer. Set the filter to as high a frequency as possible that does not affect your program. For example, try 35 Hz for music and 75 Hz for speech. On each mixer input channel, set the filter frequency just below the lowest fundamental frequency of that channel's instrument.
- Note: An amplifier reproduces the signal present on the input. Therefore, if you send clipped or squared wave signals to the input, it will recreate this on the output possibly damaging your loudspeakers.

6.2 Startup Procedure

When first turning on your amplifier, follow the procedures in the Quick-Start Guide on page 4 (stereo) or page 5 (bridge-mono).

If you ever need to make any wiring or installation changes, don't forget to disconnect the power cord.



4. WARNING: Never connect the output to a power supply, battery or power main. Electrical shock may result.

For help with determining your system's optimum gain structure (signal levels) please refer to the Crown *Amplifier Application Guide*, available online at www.crownaudio.com.

6.3 Precautions

Your amplifier is protected from internal and external faults, but you should still take the following precautions for optimum performance and safety:

1. Before use, your amplifier first must be configured for proper operation, including input and output wiring hookup. Improper wiring can result in serious operating difficulties. For information on wiring and configuration, please consult the Setup section of this manual or, for advanced setup techniques, consult Crown's *Amplifier Application Guide* available online at www.crownaudio.com.



9. CAUTION – SHOCK HAZARD: Potentially lethal voltages exist at the output connectors when the amplifier is turned on and is passing a signal.

Remember: Crown is not liable for damage that results from overdriving other system components.

6 Operation

6.4 Front Panel Controls and Indicators

Many of these functions can be disabled using Blackout Mode (a selection in the Advanced Menu, Section 7.2).

A.

Cooling Vents
Front-to-rear forced airflow through foam dust filter

B.

LCD Display Screen

Integrated 4.3" color LCD with backlight, displays amplifier settings.

C.

Power Switch

Push-on/push-off switch glows green when AC power is present at the power cord and the amplifier

circuit breaker is in the "on" position.

D.

USB 2.0 Connector

Accepts a USB drive to transfer presets, device files, and update firmware from the drive to the amplifier. This knob affect the Channel 1-4 output levels. Also Menu items and adjust parameter values that are displayed on the LCD control screen.

E.

Fault Indicator

Red LED, one per channel, flashes when the amplifier output channel has stopped operating. Usually this means that the amplifier must be serviced.

F.

Thermal Indicator

Red LED, one per channel, illuminates when the user adjust the amplifier's attenuation and muting, configure the amp, set up and view error monitoring (such as temperature and load supervision), and recall DSP presets.

G.

Clip Indicator

Red LED, one per channel, illuminates when the channel's output signal reaches the onset of audible clipping. The Clip Indicator also will illuminate during Thermal Level Control (TLC) limiting. The Clip Indicator can be turned off during Blackout mode.

H.

Level Control (Encoder)

Speed-sensitive, 0.5 dB steps, range 0 to -100 dB. This knob affect the Channel 1-4 output levels. Also Menu items and adjust parameter values that are displayed on the LCD control screen.

I.

Menu/Exit Button

"Menu" enters the main menu. "Exit" leaves the menu.

J.

Prev Button

Selects the previous item in the menu.

K.

Next Button

Selects the next item in the menu.

L.

Signal Indicators

These can be disabled during Blackout mode. Three green LEDs per channel indicate the amplifier's input and output signal levels. From top to bottom the LEDs are

-10 dB: amplifier output is 10 dB below clipping.
-20 dB: amplifier output is 20 dB below clipping.
Signal: selected input signal is above -40 dBu.

M. Ready Indicator

Green LED, one per channel, illuminates when the channel is initialized and ready to produce audio output.

N. Power Indicator

Blue LED indicates amplifier has been turned on and AC power is available. The LED will flash when the AC line voltage is 15% above or below the nominal rated range. This indicator can be turned off in Blackout mode.

O. Data Indicator

Yellow LED indicates network data activity. Data indicator flashes only when the amplifier is polled for data, or is polled to see whether it is online. This indicator can be turned off in Blackout mode.

P. Bridge Mode Indicator

Yellow LED illuminates when the amplifier is set to Bridge-Mono mode.

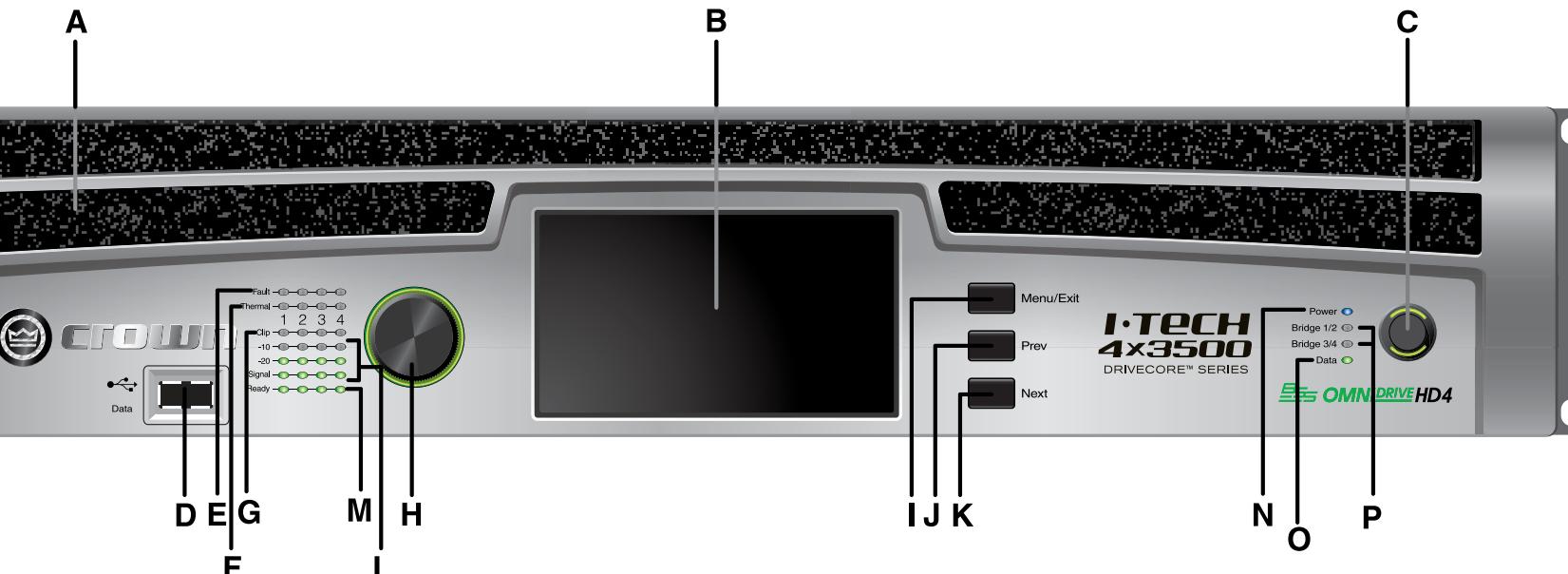


Figure 6.1 Front Panel Controls and Indicators

6 Operation

6.5 Back Panel Controls, Indicators and Connectors

A. Fans

Provide front-to-back forced airflow for cooling.

ANALOG INPUTS SECTION

B. Balanced Analog XLR Inputs

A 3-pin female XLR connector for each channel.

OUTPUTS SECTION

C. Speakon® Output Connectors

Two high-current, 50A Neutrik® Speakon® NL4MLP (mates with NL4FC and NL4), for channels 1+2 or 3+4 and one NL8 for channels 1 thru 4. Class 2 output wiring required.

These two connectors accept 2-pole or 4-pole Speakon® connectors. See Figures 2.5 through 2.10 and Tables 1, 2 and 3 for connector wiring. The top 2 Speakon connectors are wired each for stereo

operation off of one connector, the bottom NL8 speakon connection is for four way operation off of one connector.

D. Binding Post Output Jacks

One pair per channel of high-current, 60A color-coded binding posts. Accepts banana plugs, wire or spade lugs.

E. Power Cord Connector

32 amp PowerCon® AC Inlet. Voltage range is indicated above IEC inlet.

F. Reset Switch/Circuit Breaker

If the current draw of the amplifier exceeds safe limits, this breaker automatically disconnects the power supply from the AC mains. The switch resets the circuit breaker.

DIGITAL INPUTS SECTION

G. MAC Address

This specific amplifier's network identifier that is burned into its firmware.

H. Preset Indicator

LED flashes to signal the number of the current preset if active. LED is green if the preset values have not been changed once loaded. LED is yellow if the preset values have been changed since they were loaded.

I. Reset Button

To restore factory default settings: Press the Reset Button with a thin, non-conductive object, then turn on amplifier power, and continue to hold the reset button until the Preset light comes on green (approximately 16 seconds)..

J. Data Indicator

Flashes yellow only when the amplifier is polled for data, or is polled to see whether it is online.

K. AES/EBU Digital Input

This 3-pin female XLR connector accepts a digital signal in the AES/EBU format.

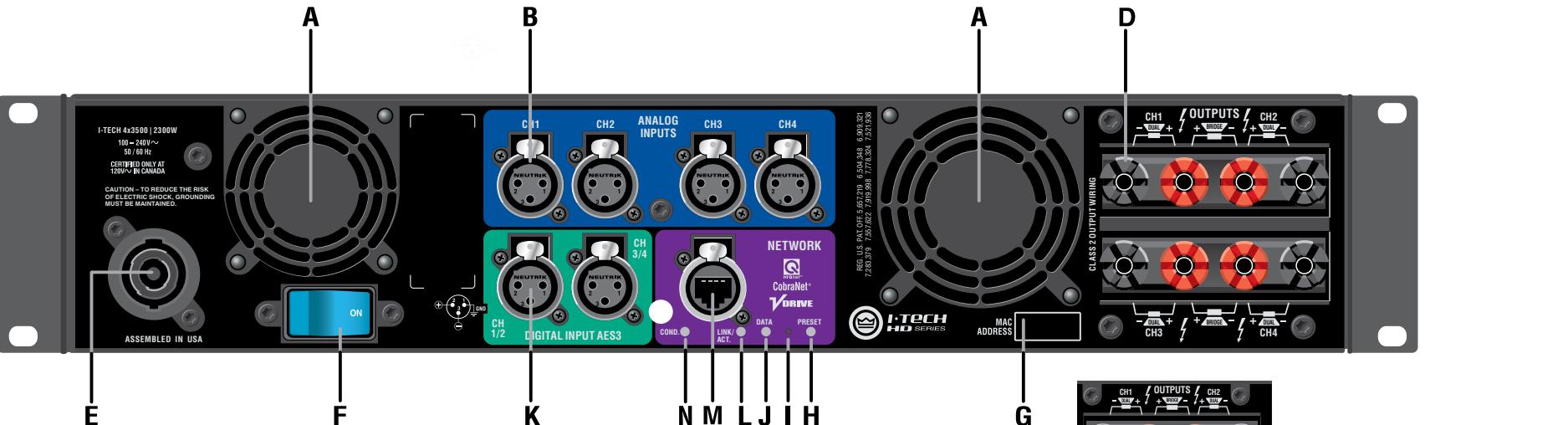


Figure 6.2 Back Panel Controls and Connectors

7 Advanced Operation

7.1 Introduction

The LCD Display Screen and its controls let you configure the amplifier and access many features that before were available only through a remote computer. Also, you can recall DSP presets via the front panel. (Some DSP parameters cannot be adjusted with the LCD Display Screen. That is done in System Architect.)

Figure 7.1 shows the parts of the LCD Display Screen. Its functions are described below. NOTE: Listed functions can also be controlled in System Architect. Here's how to access the various menus and settings in the LCD Display Screen:

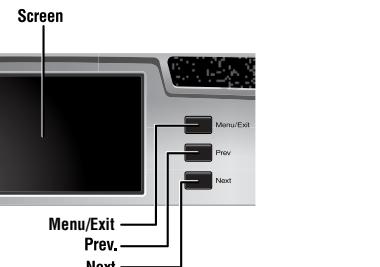


Figure 7.1 Parts of the LCD Control Screen

- Starting from the Attenuation screen, press **Menu/Exit** to go to the Menu screen.
- Press **Next** to go to the next item in the menu.
- Press **Prev** to go to the previous item in the menu.
- Turn or press the **Encoder** knob to change the value of a displayed parameter.
- When you see a menu screen, push the knob once to see the items in that menu. Or press **Next** to go to the next menu.
- Press **Menu/Exit** to leave the menu and return to the previous screen at any time.

Some menu items require confirmation: after you request a change, the display might say "Press and hold." To confirm a change, press and hold the **Encoder** knob. If you don't want the change to occur during a confirmation, turn the knob or wait five seconds.

The entire front panel or any selected screens can be locked out or set to read only status using System Architect software. Locked-out screens will either say "Lockout" or the individual parameter will say "N/A". If a change is attempted the screen will say "Changes Disabled".

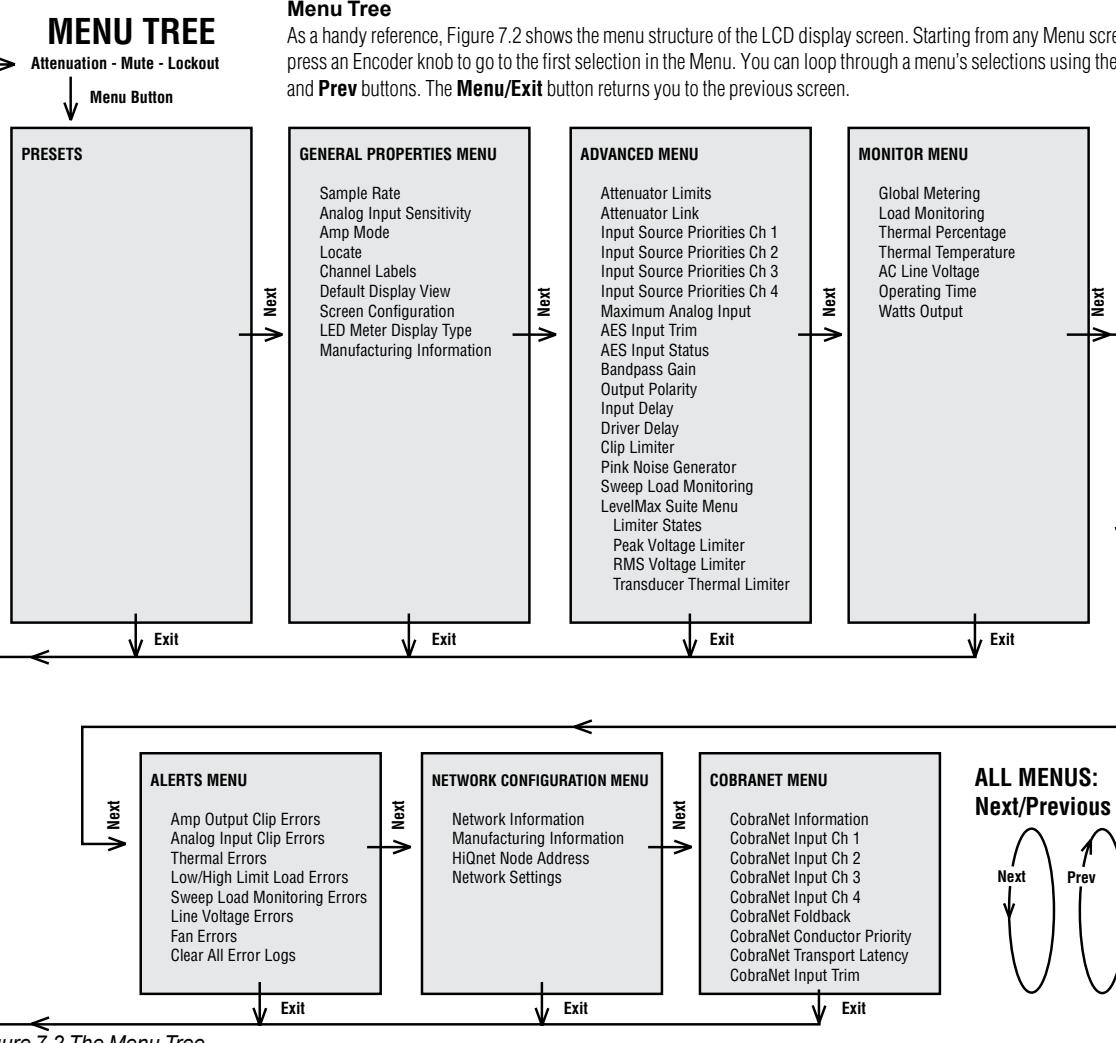


Figure 7.2 The Menu Tree

7 Advanced Operation

7.2 Operation Examples

Operation Example 1

How to set the CH1 input sensitivity using the LCD Display Screen:

1. After power-up, when the default display screen appears, press **Menu/Exit**.
2. Press **Next** twice to reach the Advanced Menu, press the encoder wheel to select.
3. Press **Next** once to reach the Analog Input Sensitivity menu, then press the encoder wheel to select.
4. Press **Next** to reach the desired channel then press the encoder wheel to select a channel and spin the encoder wheel to adjust the dB gain for that channel, press **Menu/Exit** to deselect.
5. When finished, press and hold the **Menu/Exit** button to return to the default display view.

Operation Example 2

How to monitor the Thermal Errors using the LCD Display Screen:

1. After power-up, when the default display screen appears, press **Menu/Exit**.
2. Press **Next** four times to reach the Alerts Menu, press the encoder wheel to select.
3. Press **Next** two times to reach the Thermal Errors menu, then press the encoder wheel to select.
4. Press **Next** to reach a desired channel then press the encoder wheel to clear errors on that channel, press **Menu/Exit** to deselect.
5. When finished, press and hold the **Menu/Exit** button to return to the default display view.

Now let's look at the function of each menu selection.

7.2.1 Basic Selections

Opening Screen: On power-up, the LCD Display Screen displays the Crown logo. After a few seconds, the firmware version appears, then the Attenuation screen appears.



Attenuation and Bar Meters

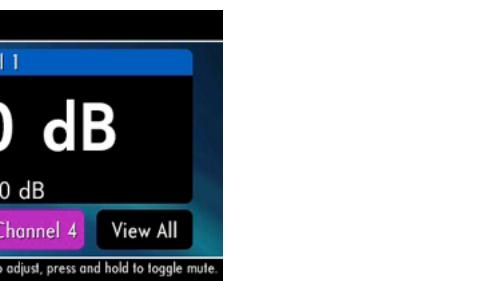
This screen displays the attenuation in dB. To change attenuation press **Next** to reach the desired channel then press the encoder wheel to select that channel and turn the Encoder. Attenuation changes 0.5 dB per detent when you turn the Encoder slowly, and more when you turn the Encoder quickly. The current preset name is displayed in the upper-right corner. The (M) is explained on the next page under Presets.

If the attenuators for both channels are linked (in the Attenuator Link screen, described later), the word **LINKED** appears. (If there is an error in the error log, an exclamation point appears at bottom center.)



Mute/Unmute (in the Attenuation screen)

- To mute a channel: The display will alternate between MUTE and the attenuation level. While the amplifier is muted, you can adjust attenuation for each channel by using **Next** and pressing the encoder wheel to select and turning the Encoder knob.
- To unmute either channel: Press and hold the encoder wheel for 1 second.



Front Panel Lockout (in the Attenuation screen)

The amplifier's administrator can control access to menu mode and the amplifier's attenuators. This lockout can be done either by the front panel buttons or by System Architect software.

To lock or unlock the menu with the front-panel buttons: press **Next** and **Prev** simultaneously. To lock or unlock the menu with System Architect software: go to the Tools Menu > Display Screen Security and press the lockout button. If a specific parameter is protected, the screen says "N/A" or "Change Disabled."

If the lockout is done from the software, the user can disable the lockout only in the software, not by pressing the front-panel buttons.

If the menu is locked and you press any button or turn an Encoder, the word "LOCKOUT" is displayed. If a specific parameter is protected, the screen says "N/A" or "Change Disabled."

To optimize the system gain structure, see Section 3.1 in the I-Tech application Guide at www.crownaudio.com/pdf/amps/137327.pdf.



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Presets: A preset is a group of DSP and amplifier settings for a particular speaker system. Turn an Encoder to view Presets 1 through 50. To recall a preset, press the Encoder, then press and hold the Encoder to confirm. For more information on presets, see Section 7.3.

The preset currently in use is displayed in the upper right corner.

(A) Active means that the amplifier is operating exactly according to that preset.

(M) Modified means that the amplifier is operating according to the preset but with some settings changed.

If a preset number is flashing, it is not in use.

In the example shown below,

- Preset 2 is in use.
- The amplifier settings have been modified (M) since Preset 2 was loaded.

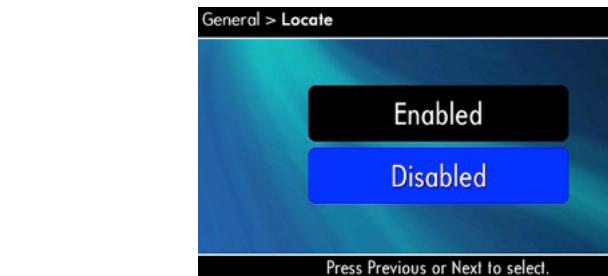


Amp Mode: This screen lets you set up the amplifier for Dual, Input Y or Bridge Mono mode. Turn an Encoder knob to choose among those three options, then press the knob to save your choice. You will be prompted to confirm your choice. The LCD screen displays the currently selected mode.

- In Dual (or Stereo) mode, both channels operate independently.
- In Bridge Mono mode, pairs of channels are summed to double the power, and the amplifier puts out a mono signal.



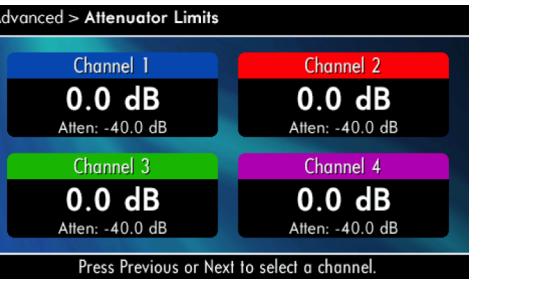
Locate: Allows users to identify a specific amplifier in a large network of amplifiers without having to compare address information. Locate can be turned on/off from software and/or the front panel. Press an Encoder knob to turn on the Locate function. When on, the LCD screen flashes. Press the knob again to turn off the Locate function. In System Architect, the device icon has a blue outline to indicate that it is in Locate mode.



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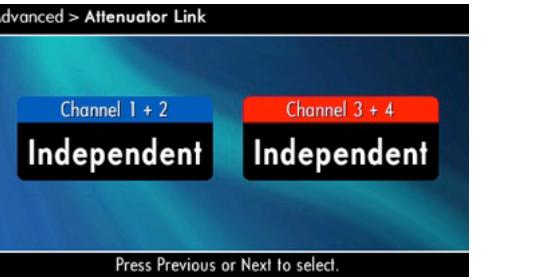
Attenuator Limits: You can set the maximum attenuation from 0 dB to -100 dB. This feature allows you to set a limit on the attenuators. Once set, attenuation cannot be adjusted beyond this level.

NOTE: The attenuator setting must be below the attenuator limit that you are trying to set. For example, if the attenuator is set at -3 dB, you cannot set the attenuator limit below -3 dB unless you decrease the attenuator level.



Press Previous or Next to select a channel.

Attenuator Link You can set the attenuators to be independent or linked. Turn an Encoder knob to choose one of those options, then press the knob to save your choice.



Press Previous or Next to select.

Input Sources: For each channel, turn the Encoder to select analog, digital, or CobraNet Options are:

- Analog
- Digital
- Digital with analog backup
- Digital with analog override

Digital with analog backup: The I-Tech HD is being fed a digital signal and an analog signal. The input is currently switched to the digital signal. If it fails, the I-Tech HD switches instantly to the analog signal.

Digital with analog override: The input is switched to the digital signal, and no analog signal is applied. If an analog signal is sent, the I-Tech HD switches instantly to the analog signal. If the analog signal fails, the I-Tech HD switches to the digital signal after a delay set by the Hold Time slider in the Input Section of the System Architect page.



Press Previous or Next to select a channel.

Maximum Analog Input: Use **Next** and press the encoder wheel to select: +21 dBu or +15 dBu. Note: Changing this value changes the range of sensitivities available to the amplifier.

For more information see Figure 7.3 and the I-Tech Sensitivity Charts in the Appendix of the I-Tech Application Guide. It is available online at www.crownaudio.com/pdf/amps/137327.pdf.



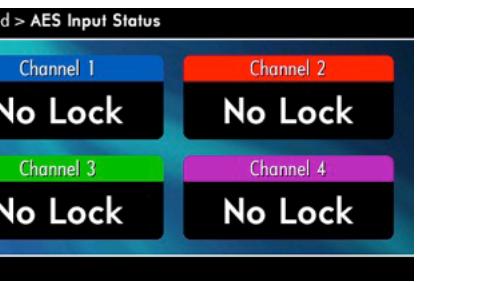
Press Previous or Next to select.

AES Input Trim: Turn the Encoder knob to vary the gain of the AES digital signal for Channel 1: -100.0 dB to +20 dB. See Figure 4.4 for more information.



Press Previous or Next to select a channel.

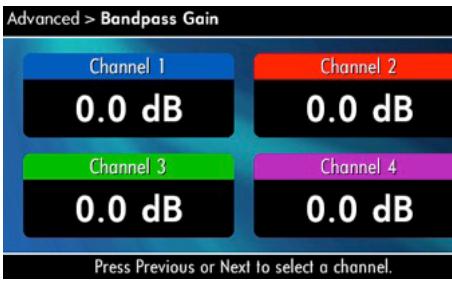
AES Input Status: "Lock" indicates that an AES cable is plugged in and the amplifier is receiving (and locking to) the AES phase logs. "No Lock" means that the amplifier is not receiving, or is not locking to the AES clock pulse.



Press Previous or Next to select a channel.

Bandpass Gain: In each channel of the I-Tech HD's DSP, just before the output limiter and after the preceding filter, is a gain block (not shown on the Signal Path block diagram). Bandpass Gain adjusts the gain of this block between -24 dB and +24 dB.

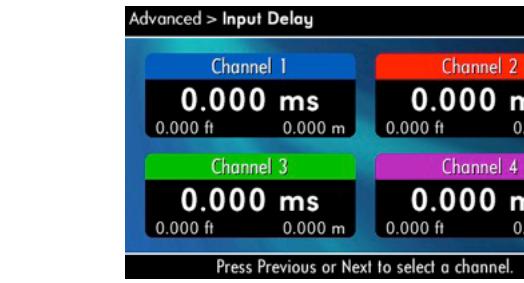
Adjusting the bandpass gain in the LCD screen makes it easy to vary the level of subwoofers, midrange drivers and high-frequency drivers.



Press Previous or Next to select a channel.

Input Delay: Sets the input signal delay in each channel. Turn the Encoder knob to vary the delay. The delay step size is speed sensitive. Pressing the encoder enables or disables the Delay.

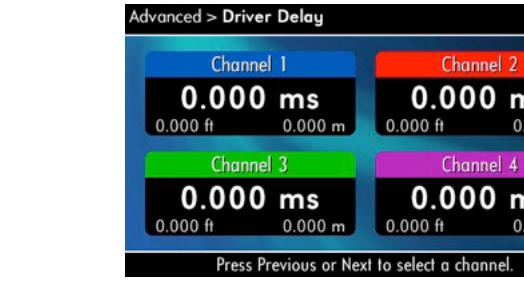
Below each channel's delay setting is the equivalent distance in feet and meters. For example, 10 ms is the signal delay of sound traveling 11.3 feet or 3.4 m.



Press Previous or Next to select a channel.

Output Delay: Sets the output signal delay in each channel. Turn the Encoder knob to vary the delay. The delay step size is speed sensitive. Pressing the encoder enables or disables the Delay.

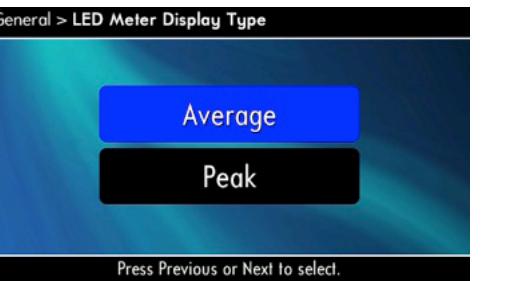
Below each channel's delay setting is the equivalent distance in feet and meters. For example, 10 ms is the signal delay of sound traveling 11.3 feet or 3.4 m.



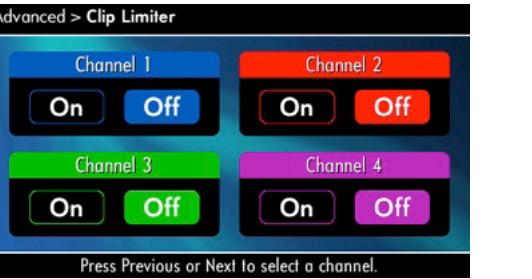
Press Previous or Next to select a channel.

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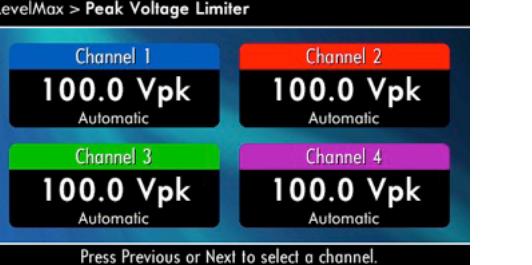
LED Meter Display Type: Here you can set the LCD bar meters to display average or peak levels. Turn the Encoder knob to select the option, then press the knob to confirm your selection.



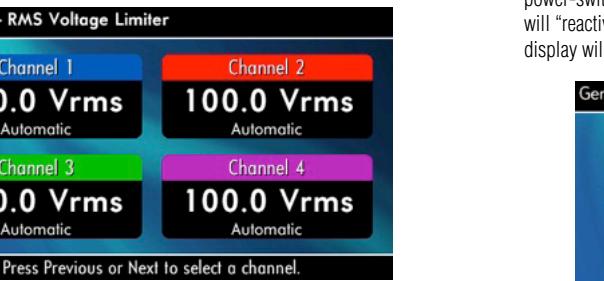
LevelMax - Clip Limiter: Limits the peak output voltage to just below clipping for each channel. Press a channel icon to turn it OFF or ON.



LevelMax- Peak Voltage Limiter: Limits the peak output voltage to a level that you set, either OFF or 1 to 500 volts, for each channel. Press an Encoder to turn on the Limiter. Once it's on, turn the Encoder to set the voltage. Additional controls, such as attack and release, are available through System Architect.



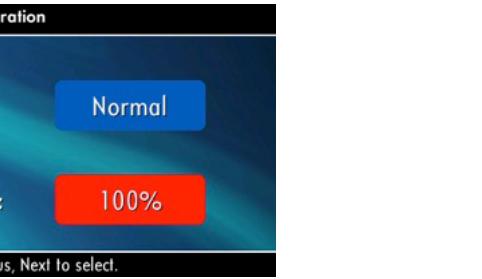
LevelMax - RMS Voltage Limiter: Limits the output rms voltage to an amount that you set, either OFF or 1 to 500 volts, for each channel. Press the Encoder to turn on the Limiter. Once it's on, turn the Encoder to set the voltage.



Pink Noise Generator: Press Next and the encoder wheel to select a channel icon to turn on the generator. Its level will read -100 dB. Adjust the noise level from -100 dB to +20 dB in 0.5 dB steps by turning the Encoder. To turn off the generator, select a channel icon or go to another menu item.

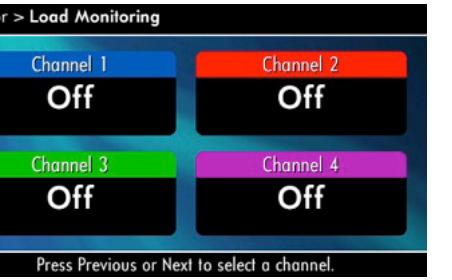


Front Panel Blackout: This screen lets you black out the front panel display unless you press a front-panel button or turn an Encoder. This feature turns off the LCD backlight and all front panel LEDs except for the fault LEDs and power-switch green LED. After blackout is enabled, an Encoder press/turn will "reactivate" the display. If no button is pressed/turned for 5 seconds, the display will return to blackout mode.

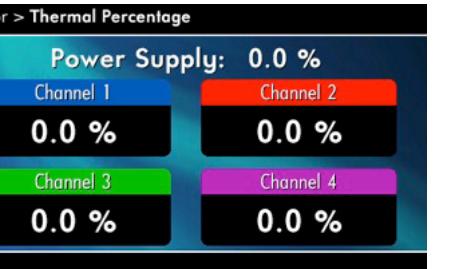


7.2.2 MONITOR MENU
This menu lets you monitor the status of the amplifier.

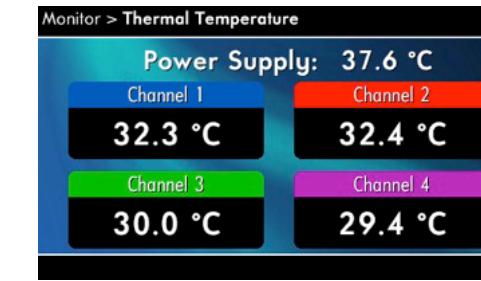
Load Monitoring: Press Next and the encoder wheel to select a channel icon to turn speaker load monitoring on or off for a channel. When load monitoring is on, the load connected to the amplifier output is displayed. The text below the reported load indicates whether the value is within the limits set in System Architect. OFF means load monitoring is off. LOW means the load is below limits. HIGH means the load is above limits. NORMAL means the load is within limits.
(Output must be sufficient)



Thermal %: This indicates the amplifier temperature in percent, where 100% is the maximum allowable temperature. If Thermal % reaches 100%, the amplifier is out of thermal headroom and has shut down (or will very soon). Measurements are shown for Channel 1, Channel 2, Channel 3, Channel 4, and PSU.



Thermal Temp deg. C: This screen displays the temperature in degrees Celsius of the Ch. 1 output devices, Ch. 2 output devices, Ch. 3 output devices, Ch. 4 output, and the power supply.

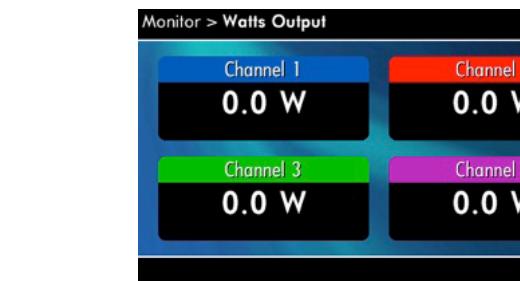


AC Voltage: This shows the AC line voltage at the AC cord inlet to the amplifier.



Operating Time: This displays the number of hours the amplifier has been on since manufacture. Like a car's odometer, it shows the total amount of operating hours and cannot be reset. All I-Tech HD amplifiers come with a variable amount of operating time on them due to burn-in and testing before shipping.

Watts Output: This displays the slow-averaged continuous output power of each channel in watts.

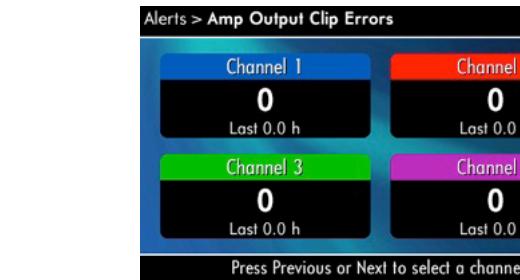


7.2.3 ALERT MENU
This menu displays a wide variety of errors in the signal, amplifier, or load.

Amp Output Clip Errors: A *clip error* occurs if the number of clip events in the amplifier output exceeds the value set with the *Count slider* (within the time set by the Time slider) on the Amplifier Settings page in System Architect.

When an error occurs, the count displayed in this screen is incremented, and the operating time since the error is displayed.

To clear this screen, select a channel icon.



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Low Limit Load Errors: This screen lets you view low-load errors without using the control software. A *low limit load error* occurs if the load impedance falls below the value set with the *Low Limit spin control* on the Amplifier Settings page in System Architect.

When an error occurs, the count displayed in this screen is incremented, and the operating time since the error is displayed.

To clear this screen, press **Next** and the encoder wheel to select a channel icon.

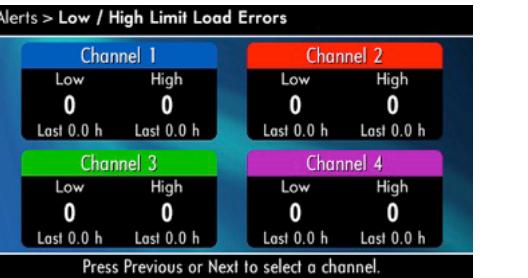
NOTE: You must use System Architect to enable error reporting and set its limits.

High Limit Load Errors: This screen lets you view high limit load errors without using the control software. A *high limit load error* occurs if the load impedance rises above the value set with the *High Limit* value on the software Control Panel.

When an error occurs, the count displayed in this screen is incremented, and the operating time since the error is displayed.

To clear this screen, press **Next** and the encoder wheel to select a channel icon.

NOTE: You must use System Architect to enable error reporting and set its limits.



SLM (Sweep Load Monitoring) Errors: This screen lets you view SLM errors without using the control software. A *low limit load error* occurs if the load impedance falls below the value set with the *Low Limit spin control* on the Amplifier Settings page in System Architect.

When an error occurs, the count displayed in this screen is incremented, and the operating time since the error is displayed.

To clear this screen, press and release an Encoder. The display will prompt you to confirm your choice by pressing and holding the Encoder.

NOTE: You must use System Architect to enable error reporting and set its limits.

Line Voltage Errors: This screen lets you view line-voltage errors without using the control software. A *line voltage error* occurs if the line voltage falls outside the values set with the *High/Low Limits slider* on the Line Voltage Page of the Control Panel.

When an error occurs, the count displayed in this screen is incremented, and the operating time since the error is displayed. The "Low" and "High" indicators show whether the line voltage was below or above the line-voltage limits.

To clear this screen, press **Next** and the encoder wheel to select a channel icon. (The display will prompt you to confirm your choice by pressing and holding the Encoder.)

NOTE: You must use System Architect to enable error reporting and set its limits.



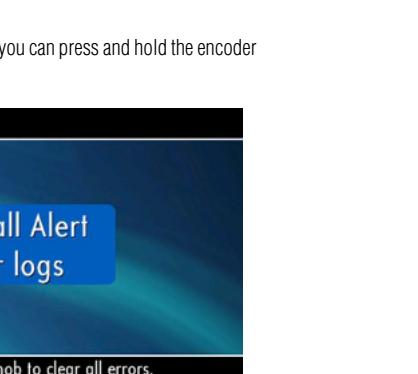
Fan Errors: When a fan error occurs, the count displayed in this screen is incremented, and the operating time since the error is displayed.

When an error occurs, the count displayed in this screen is incremented, and the operating time since the error is displayed.

To clear this screen, press and release an Encoder. The display will prompt you to confirm your choice by pressing and holding the Encoder.

NOTE: You must use System Architect to enable error reporting and set its limits.

Clear All Error Logs: In this screen, you can press and hold the encoder wheel to clear all the error logs.



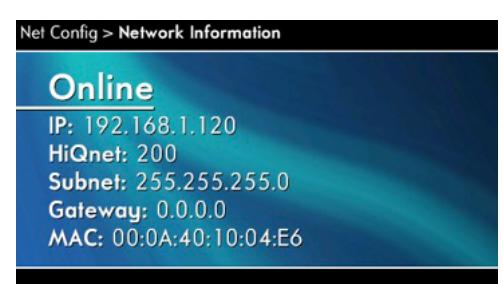
Press and hold knob to clear errors. Press Menu/Exit to exit.

7 Advanced Operation

7.2.4 NETWORKING MENU

This menu has screens for setting up audio network parameters. Push an Encoder knob to see the selections in the Networking Menu.

Network Info: This screen displays information about the audio network: IP address, HiQnet address, Subnet and MAC address. Section 12 of this manual explains these terms.



Press knob to change selected value.

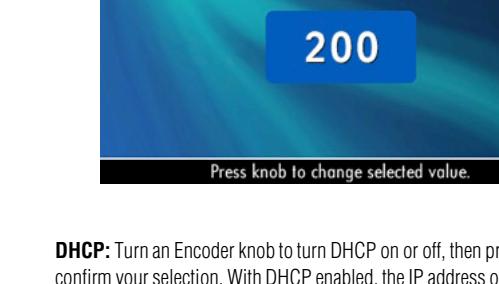
DHCP: Turn an Encoder knob to turn DHCP on or off, then press the knob to confirm your selection. With DHCP enabled, the IP address of the amplifier might change at power-up. See System Architect Help files for more information on DHCP.

IP Address: You set the IP address of the amplifier here. See Section 12 for more information on IP Address. This function is disabled when DHCP is enabled.

1. Tap the Encoder knob icon to access the left three digits, then turn the knob to set the number.
2. Tap the Encoder again to access the middle three digits, then turn the knob to set the number.
3. Tap the Encoder again to access the next digit to the right, then turn the knob to set the number.
4. Tap the Encoder again to access the right-most digit, then turn the knob to set the number.
5. Tap the Encoder again to save the IP address.

HiQnet Node Address: This screen lets you set the HiQnet address of this amplifier (a node in the network). Press the encoder wheel, then turn an Encoder knob to see the selections in the Networking Menu.

Net Config > HiQnet Node Address



DHCP: Turn an Encoder knob to turn DHCP on or off, then press the knob to confirm your selection. With DHCP enabled, the IP address of the amplifier might change at power-up. See System Architect Help files for more information on DHCP.

IP Address: You set the IP address of the amplifier here. See Section 12 for more information on IP Address. This function is disabled when DHCP is enabled.

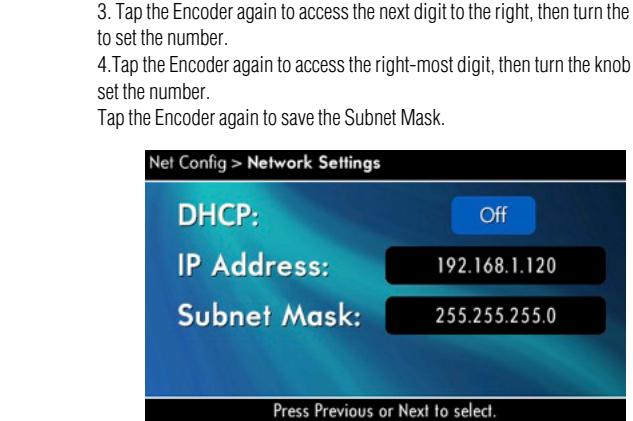
1. Tap the Encoder knob icon to access the left three digits, then turn the knob to set the number.
2. Tap the Encoder again to access the middle three digits, then turn the knob to set the number.
3. Tap the Encoder again to access the next digit to the right, then turn the knob to set the number.
4. Tap the Encoder again to access the right-most digit, then turn the knob to set the number.
5. Tap the Encoder again to save the IP address.

Subnet Mask: You set the Subnet Mask of the amplifier here. See Section 12 for more information on Subnet Mask. This function is disabled when DHCP is enabled.

1. Tap the Encoder knob icon to access the left three digits, then turn the knob to set the number.
2. Tap the Encoder again to access the middle three digits, then turn the knob to set the number.
3. Tap the Encoder again to access the next digit to the right, then turn the knob to set the number.
4. Tap the Encoder again to access the right-most digit, then turn the knob to set the number.

Tap the Encoder again to save the Subnet Mask.

Net Config > Network Settings

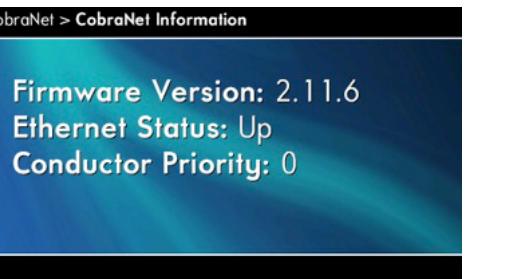


7 Advanced Operation

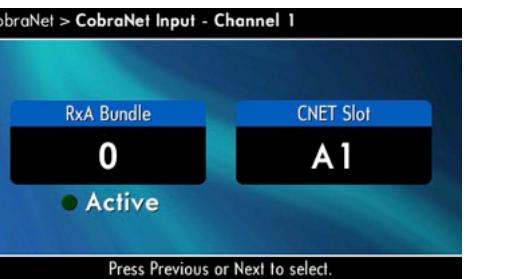
7.2.5 COBRANET MENU

To access the CobraNet Menu from the Attenuation screen, press the Encoder knob, then press **Prev**. This menu has selections for setting CobraNet parameters. Press an Encoder knob to access the CobraNet Menu selections. See Section 12 and the System Architect Help files for more on CobraNet.

CobraNet Information: This screen displays information about the CobraNet network: The firmware version, ethernet status (down or up), and conductor priority.

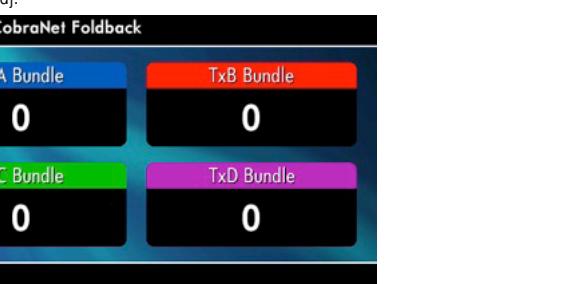


Ch1 CobraNet Rx: Tap the Encoder knob to set the number of the RxA Bundle for Channel 1. Turn the right Encoder knob to set the CobraNet input for Channel 1. The word "Active" appears if there is audio on the bundle. Bundles 1-8 are available at 48 kHz sampling rate, and bundles 1-4 are available at 96 kHz sampling rate.



Press Previous or Next to select.

CobraNet Tx: Press **Next** and the encoder wheel to select an icon, then turn the Encoder knob to adj.

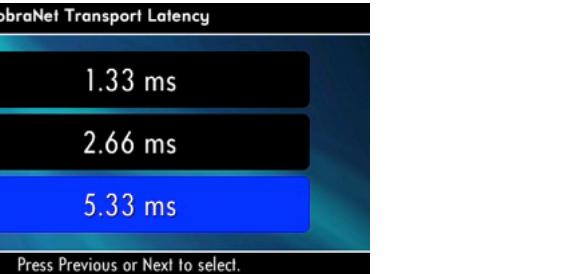


CobraNet Conductor Priority: Turn tap then turn the Encoder knob to set the priority number of the CobraNet channel. Press the knob to confirm your choice. See Section 12 for more on CobraNet conductor priority.



Turn the knob to change the value. Press knob to accept.

CobraNet Transport Latency: Press **Next** and the encoder wheel to make a selection to set the CobraNet transport latency: either 5.33, 2.66, or 1.33msec. Press the knob to confirm your choice. This latency must match the Cobranet source for proper Cobranet reception.



Press Previous or Next to select.

7 Advanced Operation

7.3 Presets

7.3.1 Introduction

Your I-Tech HD amplifier has a wide variety of onboard Digital Signal Processing (DSP). Some applications for this DSP are speaker configuration (setting the drive levels, frequency bands, delays and limiting for your particular speakers), EQ, filtering, compression, and much more. System Architect software lets you adjust the DSP settings, such as filter slope, compression ratio, EQ frequency bands, and so on.

A **preset** is a group of DSP settings that configure the amp for a specific application. For example, you might use one preset that optimizes the amp's DSP for a JBL VerTec Line Array. You might use another preset that sets up the DSP for a stereo pair of loudspeakers of your choice. You can choose any of 50 presets with the LCD Display Screen.

Preset 1 is the factory default preset and cannot be overwritten. It sets up the amplifier for stereo operation with no DSP features enabled.

The I-Tech HD amplifier works with two types of presets:

1. User presets. Using System Architect, you can create your own custom DSP presets, label them, and send them to the I-Tech HD amplifier. The amp stores those presets in firmware. You can recall those presets from the LCD Display Screen.

2. Downloadable presets. Crown and JBL engineers have designed presets that are optimized for various JBL loudspeakers, such as VerTec Line Arrays. You can download presets from www.jblpro.com. Then in the software, or with a USB drive, send the preset files to the I-Tech HD amplifier, where you can recall them from the LCD Display Screen.

We will describe each type of preset in detail.

7.3.2 User Presets

User presets are DSP presets that are configured manually. This is the basic procedure:

1. Adjust the DSP settings as desired in the System Architect software (not with the LCD Display Screen).
2. Save this group of settings as preset. Give it a label.
3. Download the preset to the I-Tech HD amplifier. See the software Help file for details. As soon as you save the setting as a preset, it is sent to the amplifier. Another option, described on the next page, is to transfer the preset file using a USB drive.
4. Select that preset from the LCD Display Screen. The preset will automatically set the DSP parameters as you set them up in the control software.

Setting some parameters of the DSP in the I-Tech HD amplifier is done using the control software, not by the amplifier's LCD Display Screen. For example, if you want to set filter Q, compression ratio, or graphic EQ, you would do so within the System Architect software.

5. When you want to recall the preset, select it from the Preset screen in the LCD display. The preset will automatically set the DSP parameters as you set them up in the control software.

7.3.3 Downloadable Presets

Crown and JBL engineers have designed I-Tech HD DSP presets that are optimized for various JBL loudspeakers, such as VerTec Line Arrays. To use them, follow this procedure in System Architect: Go to www.jblpro.com.

Section 7.8.19 describes the Presets Manager screen in System Architect.

7.3.4 Loading I-Tech HD 2 Channel Presets Into the I-Tech HD 4 Channel

In System Architect, double click on HD4. Go to File > Open Preset, find a 2 Channel ITHD preset by selecting 2 Channel Preset from file "file of type" drop down menu. Open the Preset from the and S2 will present options to load it into Channels 1/2 or 3/4.

7.4 File Transfer Via the USB Port

On the front panel of the amplifier is a USB Type A port which can transfer files either direction between a USB flash drive and the amplifier. These files can be of three types:

A Preset file, which contains amplifier DSP settings for a specific loudspeaker/amplifier configuration.

A Device file, which contain all 50 presets and the current amplifier DSP settings.

A Firmware file.

When you plug in the drive, the amplifier defaults to a file-exchange mode (and is muted). Then you can load or save individual files. No computer is needed for this file transfer.

7 Advanced Operation

7.4 List of Pop-Up Boxes and Descriptions

These messages may appear on the LCD screen.

If N/A is displayed, the particular menu has been locked out via System Architect software.

Lockout	"Lockout" is displayed when front panel lockout has been enabled from either the software or front panel. It prevents any changes from the display.
USB Lockout	"USB Lockout" is displayed when USB lockout has been enabled from the software. It disables all USB functions.
Invalid Subnet	An invalid subnet mask was selected.
Invalid Address	An invalid IP address was selected.
Preset Loaded	The selected preset has been successfully loaded.
Address Changed	The selected address has been modified. Applies to HiNet node address, IP address and subnet mask.
DHCP Changed	The current state of DHCP has been changed to either ON or OFF.
USB Fault Overcurrent!	The USB drive is drawing more than 500mA of current.
Device File is not valid!	The selected USB device file cannot be loaded because it is not valid. This is most likely a device file formatting problem. Try to create a new device file.
Loading File FAILED!	Loading the preset or device file from the USB device failed. This could be due to the format or contents of the file.
Loading File Successful!	Loading the preset or device file from the USB device was successful.
Confirm Change (Press knob again)	Displayed after the first button press. To continue, the user must release the button and press it again.
Confirm Change (Press and Hold)	Displayed after the first button press has been released. To continue, the user must press the button again.
Press and Hold (Hold knob)	Displayed after the second button press. To continue, the user must continue to hold the button for at least 2 seconds.
Change cancelled	To modify this parameter, the encoder button must be pressed twice with the second press being held down for at least 2 seconds. This message indicates that the encoder button was not properly pressed and/or held down.
No Change Selected	The selected object to change has not been modified from the current value.
Changes Disabled	The display does not have permission to change this parameter. The display observes HiNet access rights role 4. Individual screens' parameters can be user defined as Write Protected and/or Read Protected. When a parameter is Read Protected, a "N/A" is displayed.
Change Failed!	The selected change was not made. It may have been out of range or invalid.

7 Advanced Operation

7.5 Digital Audio Options (AES3 and VDrive)

Digital audio inputs allow you to keep the amplifier input signals in the digital domain. Keeping the input signal digital reduces the number of Digital-to-Analog and Analog-to-Digital conversions. This provides better sound quality and reduces pickup of electrical interference.

The AES3 connector provides a widely accepted format of digital audio transmission. Connect an AES3 signal to the AES3 connector on the rear panel. The amplifier has a digital buffer converter so it will adapt to any AES sample rate between 32 and 96kHz that is sent to it.

VDrive is a digital AES3 transport mechanism that combines a standard 10/100MB network connection with AES3 signals. VDrive employs the unused pins of a 10/100MB network to transport AES3 digital audio signals. As indicated in the chart below, one is able to transmit 4 channels of digital audio using a standard CAT5 cable.

7.6 Networking the Amplifier

If you need help understanding network concepts, please see Appendix A on Network and CobraNet Basics. Please check the System Architect help file for directions on how to use its networking configuration tools.

An I-Tech HD amplifier may be used in an existing I-Tech network and can use existing I-Tech device or venue files.

7.6.1 Network Troubleshooter

The network troubleshooter can assist you in setting up your HiNet network for the first time. Using the troubleshooter, you can address your components and be informed of addressing and other errors in the system. Please note that this wizard is designed to work with devices that are on the same physical network segment as the computer it is running on. It will not work through a router.

Select Network Card

The first page of the wizard lists all the network adapters currently in the computer. If you have more than one adapter, you can scroll through the list and see the **IP Address** assigned to each card.

If you have a card with an IP address of 0.0.0.0, it typically means one of several things:

1. The card is disabled.
2. The card is not connected to a network.

The next several pages explain some of the available DSP functions. See the System Architect Help files for more detailed information.

The VDrive digital audio transport featured in the IT4x3500HD can be enabled or disabled via the software or from the front display. This selects between AES3 through the EtherCon connector utilizing VDrive. Once AES3 has been selected either through XLR or VDrive, the preset must be stored. This selection is preset based and not a global selection for all presets loaded in the amplifier. To select AES3 between the XLR connections or VDrive, open the IT4x3500HD device panel, select source, then select between XLR or VDrive.

RJ45 Pin #	Wire Color (T568A)	Wire Diagram (T568A)	10Base-T Signal 100Base-TX Signal	1000Base-T Signal
1	White/Green		Transmit +	BI_DA+
2	Green		Transmit -	BI_DA-
3	White/Orange		Receive +	BI_DB+
4	Blue		AES 1/2 +	BI_DC+
5	White/Blue		AES 1/2 -	BI_DC-
6	Orange		Receive -	BI_DB-
7	White/Brown		AES 3/4 +	BI_DD+
8	Brown		AES 3/4 -	BI_DD-

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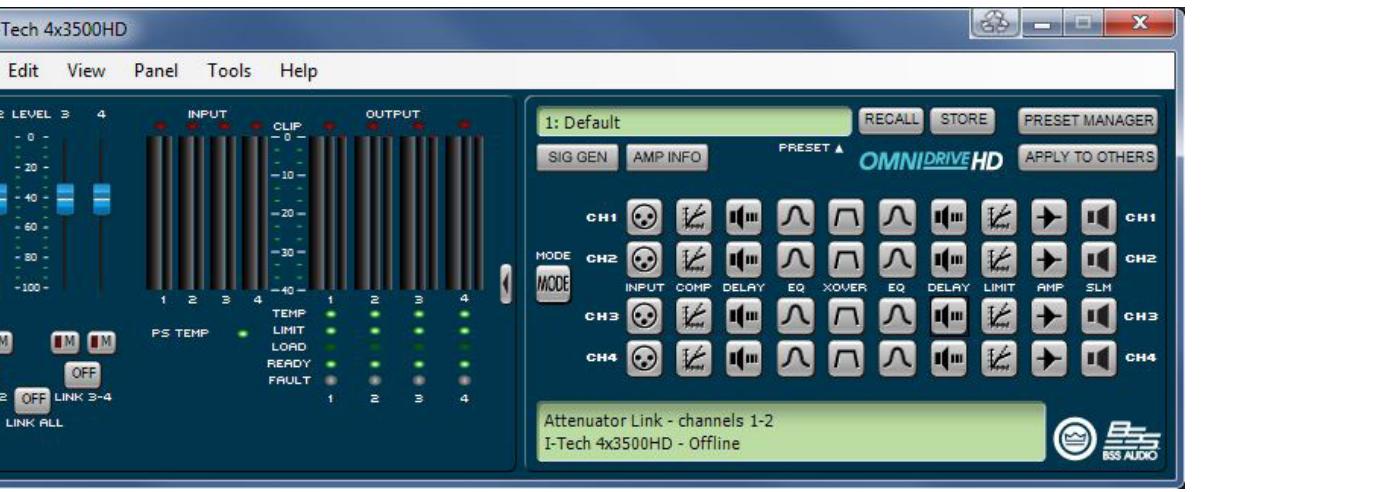
7.7.1 Amplifier DSP Selection Screen

On the left of this screen is the Input and Output Signal Level Monitor (explained later on this page). On the right side are rows of DSP selection buttons for each channel. When you double click on a button, a control panel opens for that button's function.

Also on the right are buttons that open control panels for these functions:

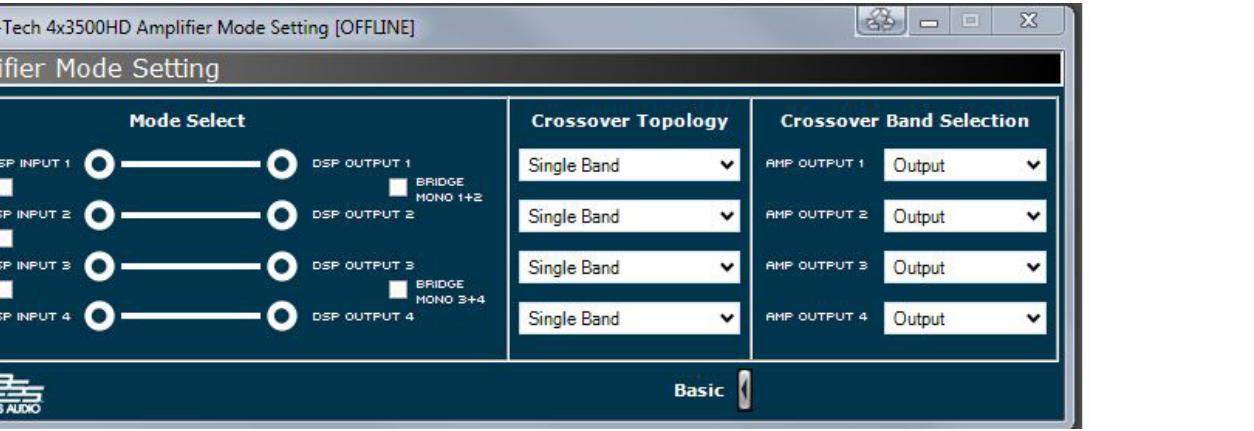
- Amplifier mode (dual, bridge-mono, input Y)
- Signal generator (pink noise and sine wave)
- Amplifier information
- Recall and store presets
- Preset manager
- Apply to others (apply this amplifier's DSP settings to other devices)

We'll explain each of these functions on the next several pages.



7.7.2 Amplifier Mode

This is accessed by double clicking the mode button.



7 Advanced Operation

7.7.3 Input Level, Fades, Mute, Link, I/O Level Meters, and Indicators

This panel is on the left side of the I-Tech HD's main control panel shown above. Channels 1, 2, 3, and 4 Level Controls set the input signal level of each channel. Each channel can be muted, and both faders can be linked with the Link button.

Input Signal Level Meter (green)

The measurement range is from 0dBFS to -40dBFS with 0.5dB resolution. Each audio channel has the following monitors:

- **Switched Audio Input:** This meter shows only the switched (selected) audio input after the Input Audio Router.
- **Clip Indicator:**

Output Signal Level Meter (blue)

The measurement range is from 0 dB to -40 dB with 0.5 dB resolution. The monitors are scaled so that 0 dB is referenced to the full rated output voltage of the amplifier. The wider left meter is rms and the right meter is peak for each channel. An output clip indicator is provided.

Indicators

Temp: Lights if the temperature of the amplifier is too high and the amplifier is near shutdown. Green: Thermal headroom used up is less than 59%.

Yellow: Thermal headroom used up is 60-80%.

Red: Thermal headroom used up is more than 81%.

Limit: Lights when the amplifier signal is being limited.

Load: Lights when the load impedance is out of range.

Ready: Lights when the amplifier is on and ready to supply power.

Fault: Lights when the amplifier is in a fault mode service may be required.



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7.7.4 Input Signal Router

Each channel of the I-Tech's signal processing has an Input Signal Router that lets you choose the audio signal that will be used by the channel. Choose one of the following configurations:

- Analog Audio:** This is the audio input from the audio connector on the back panel of the amp. Choices for the Analog Audio input are Channel 1, Channel 2, or a sum of Channels 1, 2, 3, 4, or a sum of 3 and 4.
- Digital Audio:** This is the audio input from the AES/EBU connectors or CobraNet connector on the back panel of the amp. It can be set up for Digital with Analog Backup or Digital with Analog Override:

Digital with Analog Backup

The input takes the digital signal as its source when the digital signal (or just the digital clock) is present. If the digital lock is lost, the input switches to the selected Analog Source. Once that digital lock is found again, the input switches back to the selected digital source. This works even when the digital signal has no audio but is sending a clock signal.

Digital with Analog Override

The input takes the digital signal as its source. If the analog signal level is greater than an Analog Signal Detect Threshold that the user defines, the input switches to the analog signal. As long as the analog signal level exceeds the threshold, or as long as its level falls below threshold for a user-defined hold time, the input remains the analog signal.

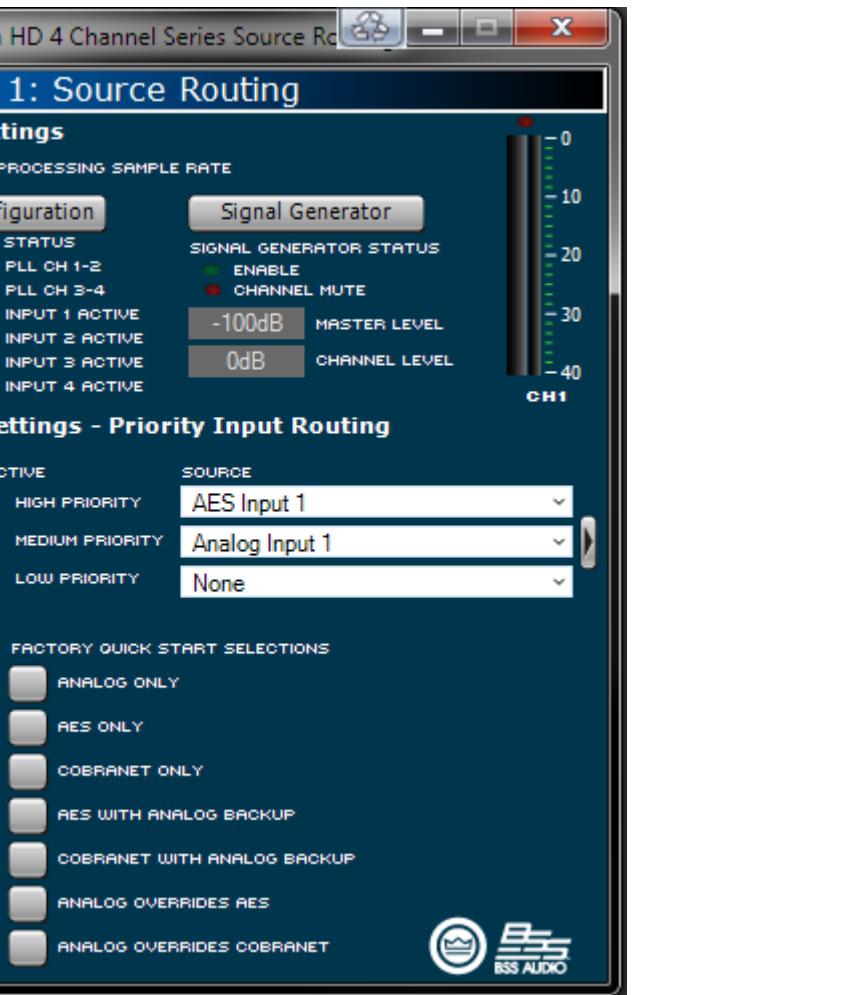
Analog Signal Detect Threshold (-100 dB to 0 dB)

This option is not displayed unless override is selected. If the selected analog signal exceeds the threshold level, the analog signal becomes the source for this input rather than the selected digital input source. Use the numeric spin box or the fader provided to set the threshold.

Hold Time (0 - 60 seconds)

This option is not displayed unless override is selected. Once an overriding analog input signal falls below threshold, the Hold Time determines how long the input will continue to be the selected analog input before reverting to the selected digital input. Use the numeric spin box or the fader provided to set this time.

The control operation will vary based on the mode that is selected.

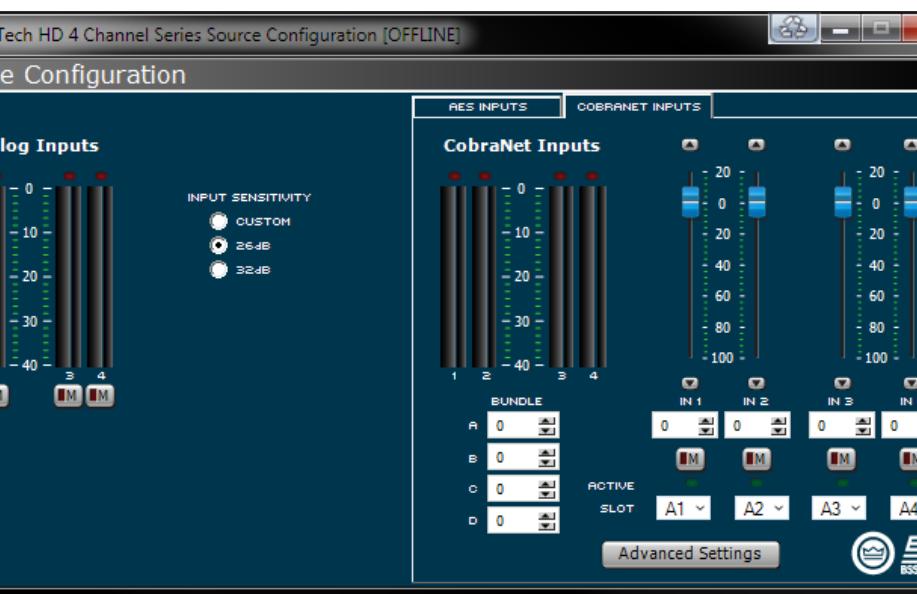
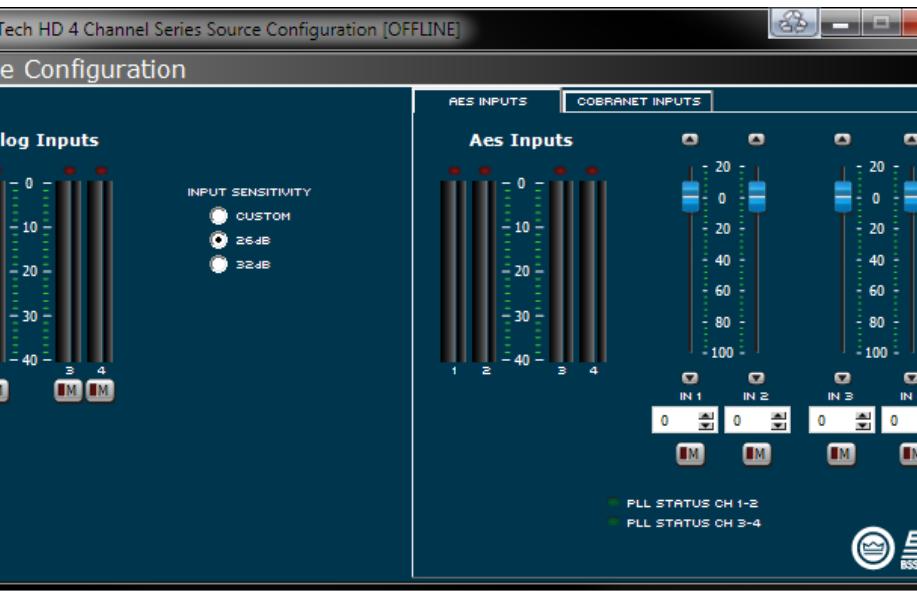


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7.7.5 Source Configuration

In this screen you can configure the levels and settings of the analog, AES and CobraNet inputs. The input signal of each channel can be independently muted.

The I-Tech HD amplifier can be set to work as a fixed gain amplifier: any input signal is given a fixed amount of gain, either 26dB or 32dB. In addition, you can set the input sensitivity to determine how much gain is provided. See Section 4.8.20 for more details.



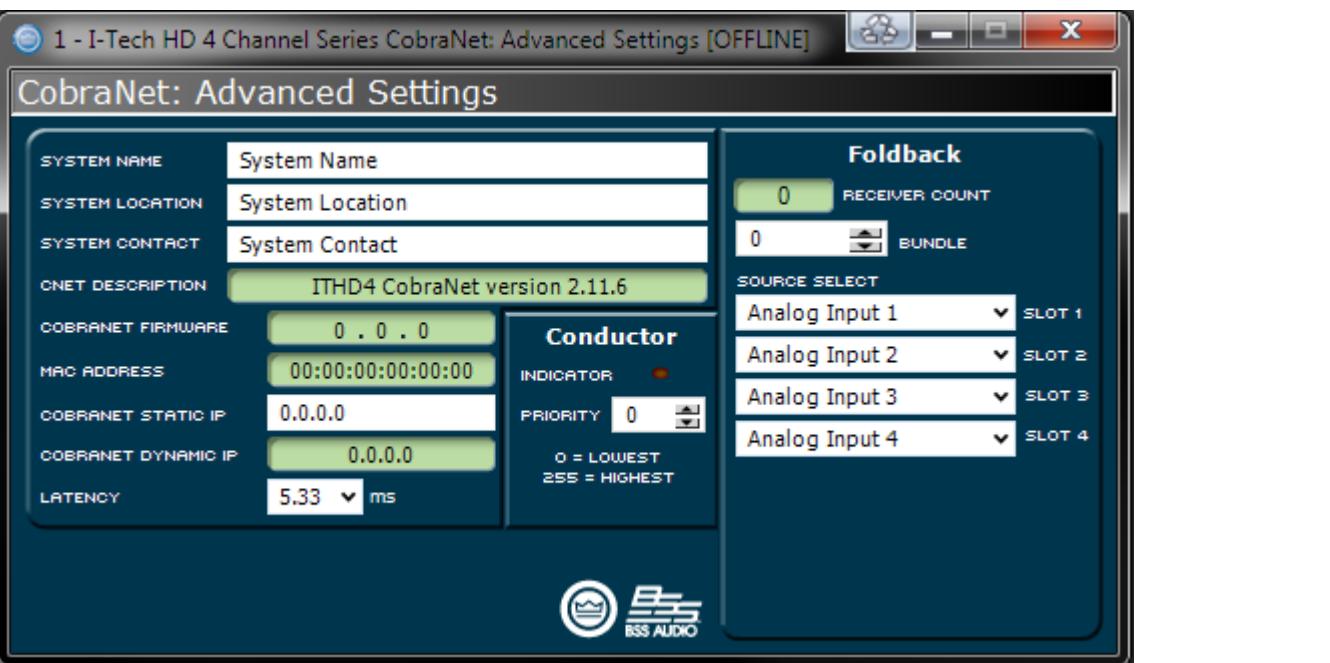
4 Advanced Operation

7.7.6 CobraNet Advanced Settings

In the CobraNet Advanced Settings screen you can set the parameters shown here. Indicators that are green can't be modified.

The latency on this screen is the CorbaNet transport latency or buffering. This latency must be set to the same latency as the source CobraNet device.

The Foldback control panel lets you select the items to transmit via CobraNet. Selections available are the Analog Inputs, Amplifier Outputs or AES input signal. The audio can be routed out through only one bundle.



7.7.7 Input Signal Compressor

An input signal compressor is available for each channel. Several parameters control this feature:

On/off button: Enables or disables this function.

Threshold: Sets the level, in dBu, above which the compressor begins to attenuate the input signal. This level corresponds to the input level meter reading. The compressor is "feed-forward," meaning that the level detection point is located before the gain control stage. The range is from +20 dBu to -50 dBu.

Attack Time: Sets the attack time of the compressor. The attack time is defined as the time it takes the compressor to attenuate the signal gain by 20 dB. The range is from 1 millisecond to 0.1 second.

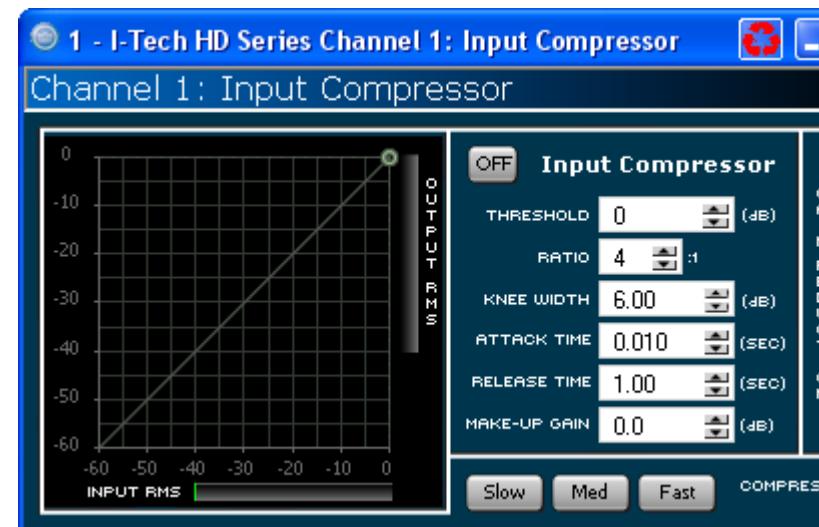
Release Time: Sets the release time of the compressor. The release time is defined as the time it takes the compressor to increase the signal gain by 20 dB. The range is from 10 milliseconds to 10 seconds.

Compression Ratio: Sets the compression ratio of the compressor. The ratio is the relationship between the amount of attenuation applied by the compressor versus the amount that the input signal is over the threshold. The available settings are 2:1 to 32:1.

Makeup gain: The amount of gain you want to apply to the compressed signal to raise it to a higher level.

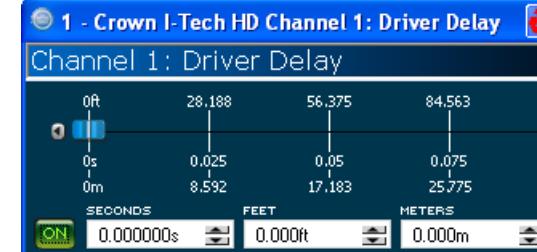
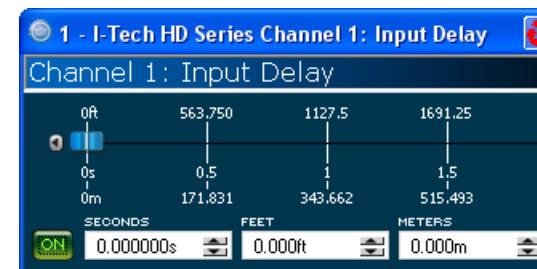
Knee width: The knee of the compression curve is the transition point between no compression and compression. The wider the knee, the more gradual is the transition -- so compression is not so obvious when the signal is close to threshold.

Beyond the controls for each channel, a single-control **Compressor Link** connects the compressors together. The compressors are tied at their sense points, meaning that the greater of the two input signals will be used as stimulus for both compressors. Each compressor will still compress based on its individual threshold, attack, release and ratio settings.

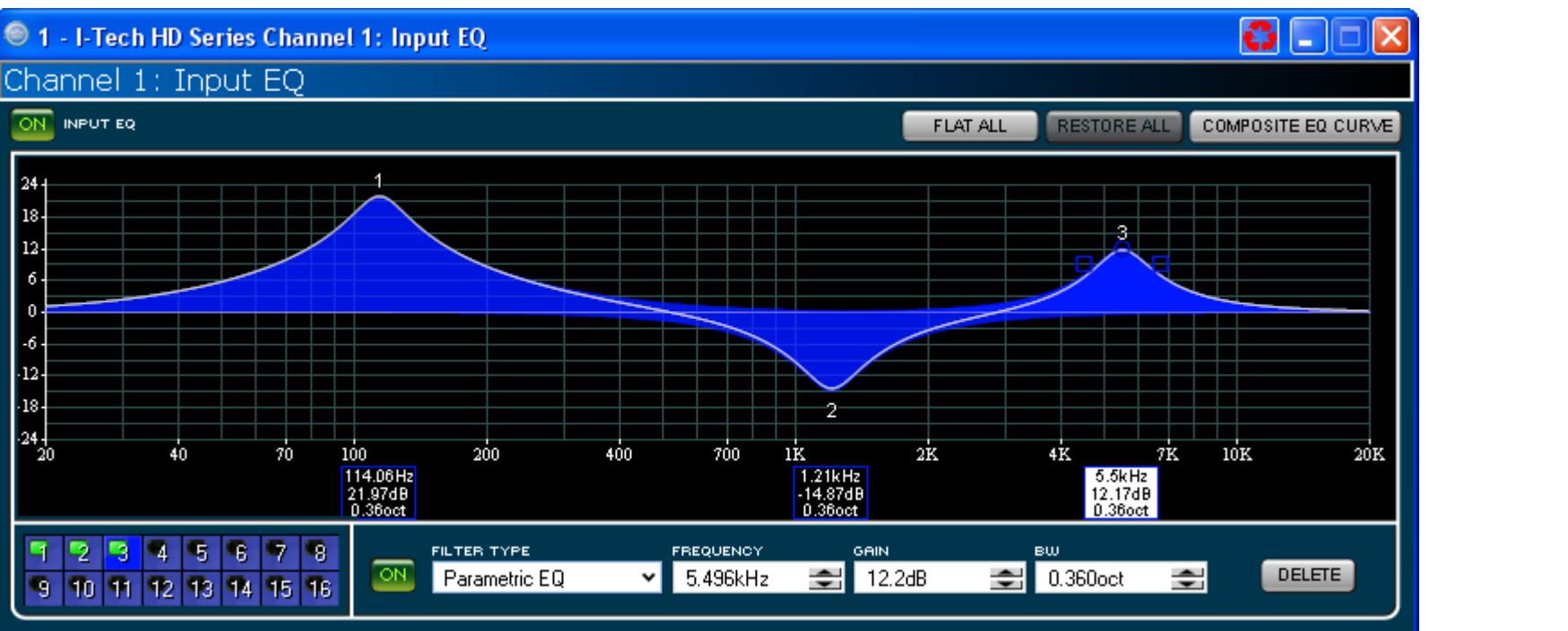


7.8.8 Input Delay and Driver/Output

In these screens you can set the signal delay for each channel in seconds, feet or meters.



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7.7.9 Input EQ and Output EQ

These screens let you adjust channel equalization for up to 16 frequencies. You can select filter type, frequency, gain, and bandwidth in octaves or Q as set by the user preferences in System Architect. Changes to the equalizer's frequency response can be done by typing in parameters or by click-dragging the response curve.

FIR advantages: FIR has a longer processing latency but has less phase artifacts compared to IIR. FIR allows for brickwall, 48dB/oct and 24dB/octave filters with zero phase shift.

IIR advantages: IIR has one processing block per channel; IIR has one or more. For example, you can place an FIR at only one end of a channel, but you can place an IIR at either end.

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7.7.10 Crossover Filters

The Crossover section lets you use infinite impulse response (IIR) or finite impulse response (FIR) filters for the crossover.

FIR advantages: FIR has a longer processing latency but has less phase artifacts compared to IIR. FIR allows for brickwall, 48dB/oct and 24dB/octave filters with zero phase shift.

IIR advantages: IIR has one processing block per channel; IIR has one or more. For example, you can place an FIR at only one end of a channel, but you can place an IIR at either end.

The amplifier contains four FIR blocks which can be used in various configurations:

In Duplicate mode the same settings are used on both channels, allowing two FIR filters per bandpass. In Unique mode the FIR filters are limited to one per bandpass. However, if the High Pass of Channel 1 and Low Pass for Channel 2 are set to identical slopes and frequencies, you can use FIR on all the bandpasses. The logic for this operation is built into the software controls. In addition, the Processing (FIR) latency sets the minimum frequency that can be set for an FIR.

A polarity inverter button lets you change the polarity of each channel independently.

Each audio channel has three separate places where filters can be placed in the system: input EQ (16 filters), Crossover, and Output EQ. Up to 24 filters per channel are available, plus crossover filters and 4 filters for the signal generator. Each filter has up to +/-24dB of gain. The following filters are available:

- Lowpass: Bessel 2-4, Butterworth1-4, and Linkwitz-Riley 8
- Highpass: Bessel 2-4, Butterworth1-4, and Linkwitz-Riley 8
- Lowshelf: Low-frequency shelving EQ
- Highshelf: High-frequency shelving EQ
- Lowpass EQ: Variable bandwidth, can be expressed as Q (0.100 to 35) or Octaves (0.041 to 6.672)
- Highpass EQ: Variable Q from 0.1 to 25
- Parametric EQ: Variable Q form 0.1 to 35

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7.7.11 LevelMax Suite

This is a suite consisting of a peak voltage limiter, average power limiter, clip limiter and thermal transducer limiter. First you will set the mode to automatic or advanced:

- Automatic mode:** The software determines the best settings based on the signal characteristics. You can modify only the RMS threshold, Speaker Thermal threshold and time constant. Everything else is set automatically.
- Advanced mode:** You can modify the RMS threshold, Speaker Thermal threshold, time constant, Peak threshold and Peak/RMS release.

Peak Voltage Limiter

This limits the peak voltage output of the amplifier.

On: Enables or disables this function. You'll set the parameters below:

- Threshold:** Sets the level, in absolute voltage, which the limiter will allow from the amplifier. The range is from $12V_{pk}$ to $255V_{pk}$.
- Attack Time:** Sets the attack time of the limiter. The attack time is defined as the time it takes the limiter to attenuate the output signal by 20 dB. The range is from 1 millisecond to 100 milliseconds. The instant setting attack time is instant (0 millisecond).
- Release Time:** Sets the release time of the compressor. The release time is defined as the time it takes the limiter to increase the output signal by 20 dB. The range is from 10 milliseconds to 10 seconds.

Average Power Limiter

This limits the long-term output power of the amplifier. You'll set the parameters below:

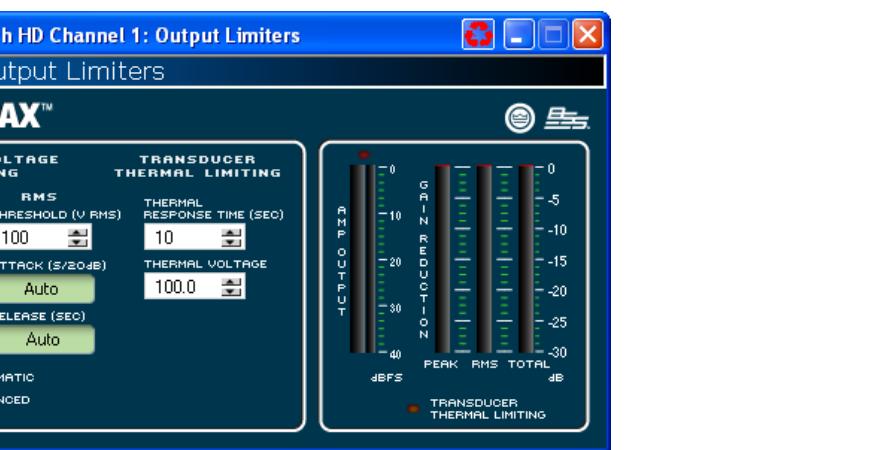
- Threshold:** Sets the average power level, in watts, which the limiter will allow from the amplifier channel. The range is from 10 watts to 10,000 watts. This level should be set to the connected loudspeaker's long-term power rating. Also you must enter the loudspeaker's nominal impedance.
- Attack Time:** Sets the attack time of the limiter. The attack time is defined as the time it takes the limiter to attenuate the output signal by 20 dB. The range is from 1 second to 30 seconds.
- Release Time:** Sets the release time of the compressor. The release time is defined as the time it takes the limiter to increase the output signal by 20 dB. The range is from 1 second to 30 seconds.

Clip Eliminator: This limiter monitors the clip events and attenuates the input signal to prevent amplifier clipping. The attack and release times are preset to optimum rates. Only one parameter controls this limiter. "Enable" enables or disables this limiter.

A lookahead feature prevents overshoot of the limiters. These settings were developed with JBL to protect the speakers from overvoltage.

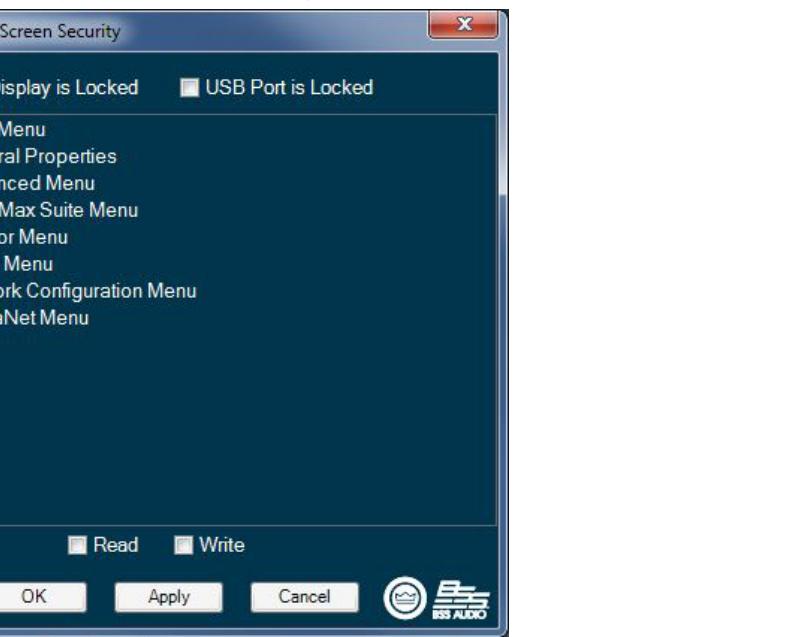
Transducer Thermal Limiting

This limits the long-term output power of the amplifier to what the loudspeaker load can handle without overheating and going into thermal compression. You can set the thermal level at which the thermal limiter engages, the thermal response time and thermal voltage. Your loudspeaker manufacturer might be able to provide this information. Meters show the amplifier output level in dBFS; and peak, rms, and total gain reduction.



7.8.12 Front Panel Security

Here you can choose which amplifier LCD screens to lock so they can't be changed.



7.7.13 Amplifier Settings

The **Amplifier Output Enable** button turns each channel on or off.

Error Reporting

The I-Tech amplifier can detect four different error conditions per channel and errors in the AC line voltage. Each error type can be individually configured to report the error through the network. Network reported errors appear in the control software Event Log. The following describes each error source.

Clip: A clip detector is provided each channel. These monitors will indicate any clip condition ($>0.05\%$ distortion) in the amplifier channel. The clip detectors also can indicate Thermal Level Control limiting.

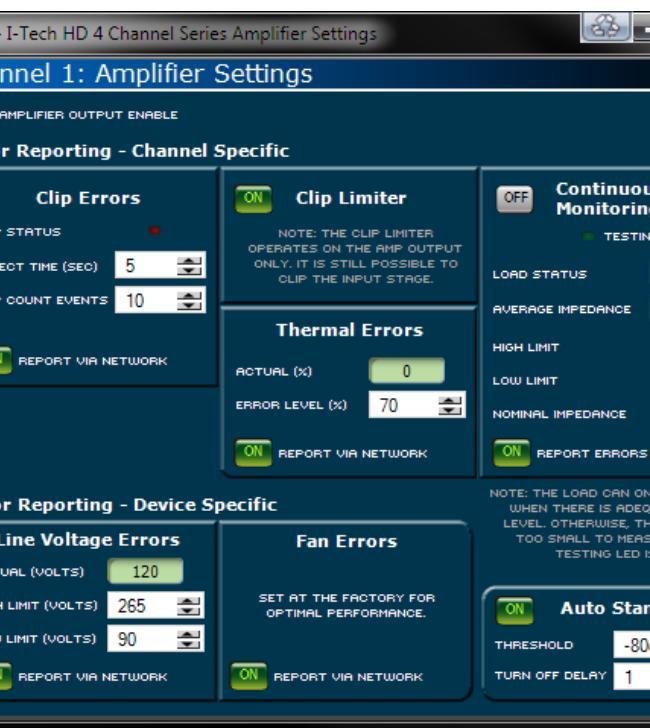
The I-Tech HD can be configured to report if an excessive number of output clip events occur in either amplifier channel. The clip events are considered an error if they exceed the defined count per the defined unit of time.

The **Count** Control sets the maximum number of events before the error is reported. The range is 1 to 100. The **Time** Control defines the amount of time that the events are counted before starting the counting process over. Its range is 1 to 10 seconds. Note: Clip events are defined as the start of each clip and every 10 milliseconds the amplifier is in clip thereafter. Using a high count setting with a short time setting may result in clip errors never being detected.

Thermal: Errors can be generated for excessive temperature in the output section of the amplifier channel. The **Threshold** Control sets the level that, if exceeded, will generate the error report. This control has a range of 1 to 100%.

Line voltage: The AC line voltage can be monitored and an error generated whenever the voltage is outside of the limits set by the software.

Fan: Using the ON button, you can enable reports of any fan errors.



Continuous Load Monitoring

The load supervision feature allows real-time monitoring of the load connected to each amplifier channel. When enabled, the I-Tech continuously monitors the amplifier output voltage and current, and calculates the long-term average load impedance. The measured load impedance is compared against the user-defined high and low limits. If either limit is exceeded, the status indicator and, if enabled, the System Architect Error Reporting functions alert the user of the problem. There are six controls and two indicators for each channel:

On: Enables or disables the load-supervision function.

Load Status: This indicator shows the present status of the load impedance with respect to the user-defined high and low limits.

Average Impedance: Shows the average impedance for the connected load.

High Limit: Sets the upper bound above which the system will report a "high" error status.

Low Limit: Sets the lower bound below which the system will report a "low" error status.

Nominal Impedance: Sets the expected average impedance for the connected load. This value determines the output signal level required for the test. This parameter is also used by the average power limiter to determine the expected power threshold. You set this field to the nominal impedance of the connected load per channel.

Report Errors via Network: When this is turned on, high and low load errors are sent via the network to System Architect software.

7 Advanced Operation

7.7.14 Amplifier Information

Information about the amplifier is stored in the software system inventory and is displayed on the software control panels. Information includes manufacturer, model, date code, and serial number. The green items can't be edited.

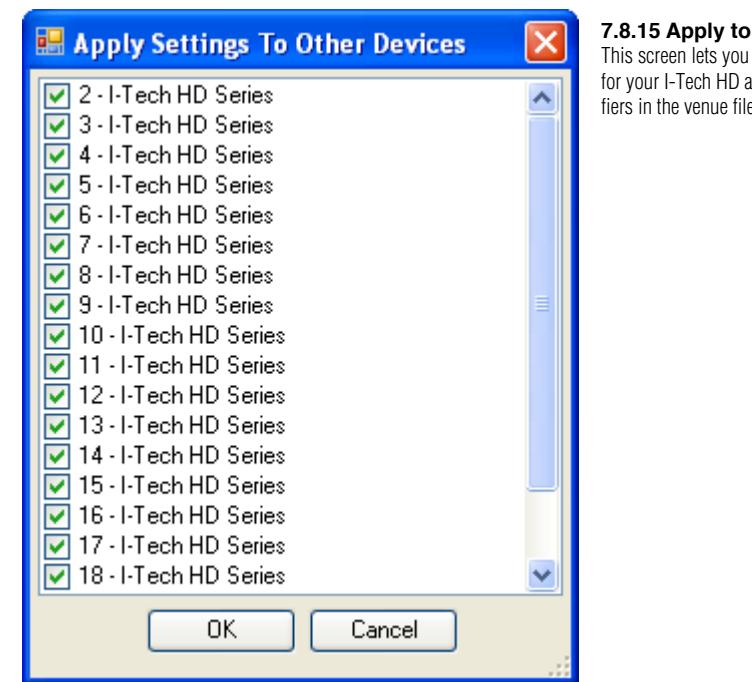
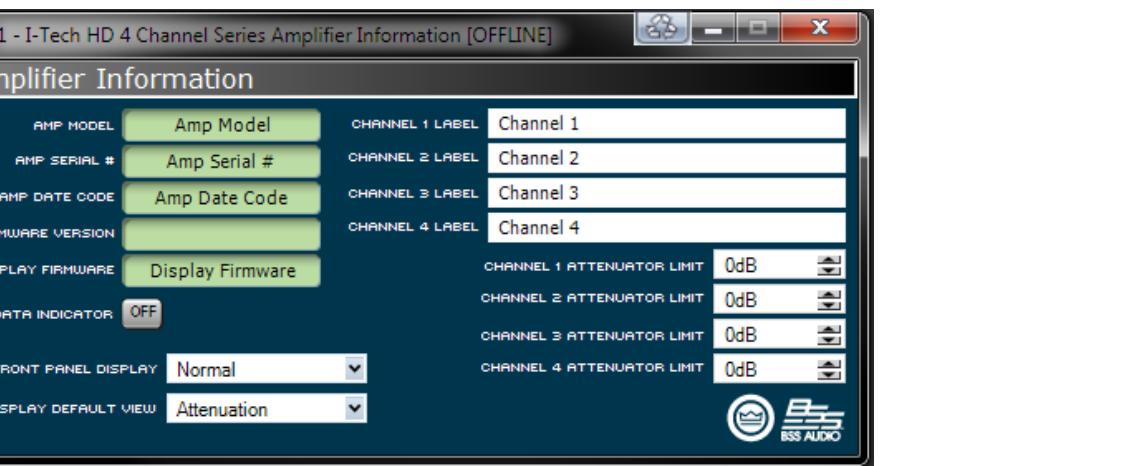
Channel Labels: Each amplifier includes two labels that you may edit for your own use. Simply type in the text areas. Use these labels to describe the channels of your amplifier. These names will appear in the individual processing blocks.

Attenuator Limits: This lets you set a limit on the attenuators. You can set the maximum attenuation from 0 dB to -100 dB. Once set, attenuation cannot be adjusted beyond this level.

NOTE: The attenuator setting must be below the attenuator limit that you are trying to set. For example, if the attenuator is set at -3 dB, you cannot set the attenuator limit below -3 dB unless you decrease the attenuator level.

Front Panel Display: This button lets you blackout the front panel display until you press a front-panel button or turn an Encoder. This feature turns off the LCD backlight and all front panel LEDs except for the fault LEDs and power-switch green LED. After blackout mode is enabled, an Encoder press/turn will "reactivate" the display. If no button is pressed/turned for over 5 seconds, the display will return to blackout mode.

Front Panel Display:



7.8.15 Apply to Others

This screen lets you apply the System Architect settings for your I-Tech HD amplifier to other I-Tech HD amplifiers in the venue file.

7.8.16 Delay (Latency)

Due to the nature of DSP processing, there is some inherent delay or latency within the I-Tech HD's DSP. These latencies are shown in Table 3 below:

Table 3 I-Tech HD latencies at two sample rates

Sample rate	ADC	DSP	DAC	Amplifier	Total latency
48 kHz	250µs	666µs	196µs	54µs	1.166ms
96 kHz	94µs	333µs	48µs	54µs	1.13ms

These latencies do not include CobraNet Transport Latency and FIR Processing Latency.

7 Advanced Operation

7.7.17 Preset Manager

The control settings for all the functions can be stored as presets. Up to 50 presets can be saved in the I-Tech amp's nonvolatile memory. Each preset can be named with a 30 character name describing its function, which is also stored in memory.

- The **Store Preset Control** saves the current setting to the specified preset.
- The **Recall Preset Control** recalls the specified preset settings. It appears in the Active Preset field. If you change the settings after recalling a preset, the Store button will turn yellow.

You can drag the presets to place them in a logical order. Also, you can color-code the presets to organize them.



7 Advanced Operation

7.7.18 Signal Generator

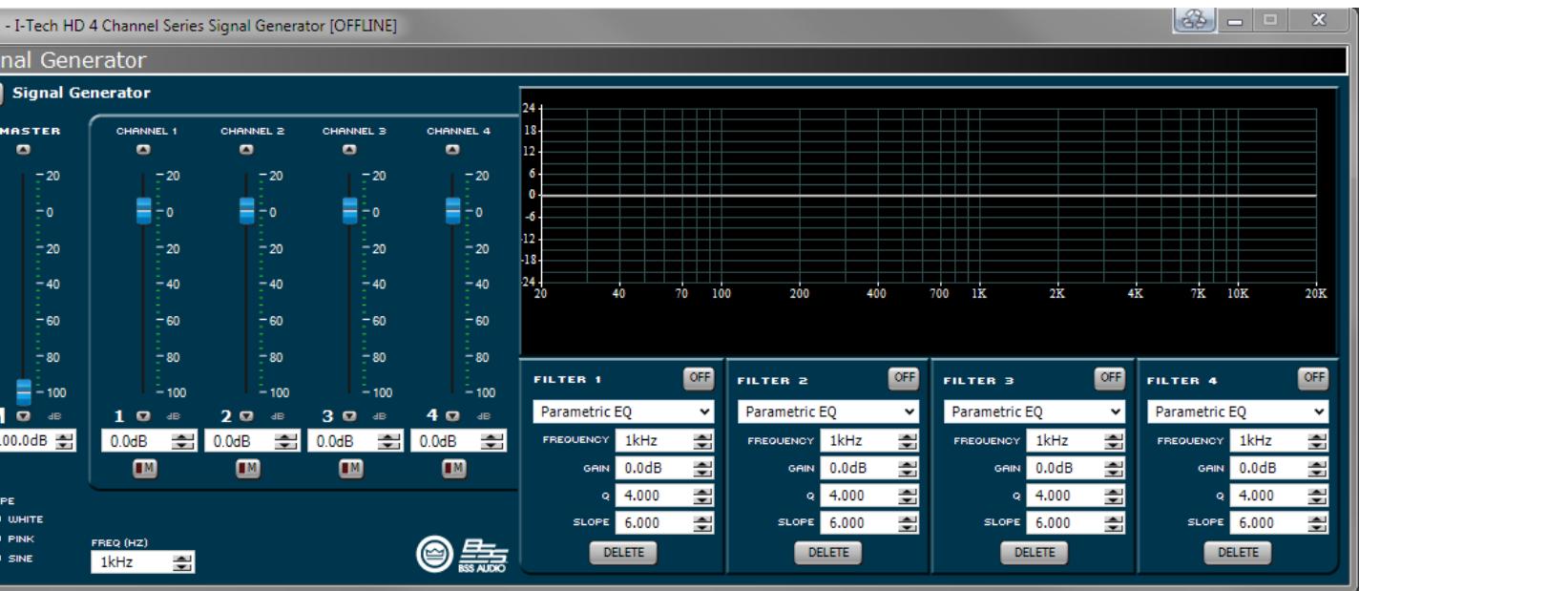
All channels share a **noise generator** that allows noise to be mixed into the audio signal. This is useful for noise masking applications and testing. Each channel has the following controls:

- **Noise On/Off:** Turns the noise generator on or off.
- **Noise Type:** Selects full-spectrum white noise or pink noise.
- **Master:** Sets the noise level with a fader.

The **sine-wave signal generator** lets you mix a single tone into the audio signal. Its controls are listed below:

- **Sine Frequency:** Sets the frequency anywhere from 20Hz to 20kHz.

The noise generator and sine-wave generator cannot be used at the same time.



7 Advanced Operation

7.7.19 Fixed-Gain Mode in the I-Tech HD

Fixed-gain mode makes any I-Tech HD model have the same gain, regardless of output power.

To do that, fixed-gain mode sets the **Analog Input Sensitivity** to 0 dB gain, then adjusts the **Fixed Gain Compensation** fader, and the **Maximum Analog Input**, to achieve 26 or 32 dB of gain (if the Maximum Analog Input is set Low), or to achieve 32 or 38 dB of gain (if the Maximum Analog Input is set High), no matter what model the amplifier is.

However, In fixed-gain mode, the input trims (Analog, AES, CobraNet) can still be adjusted.

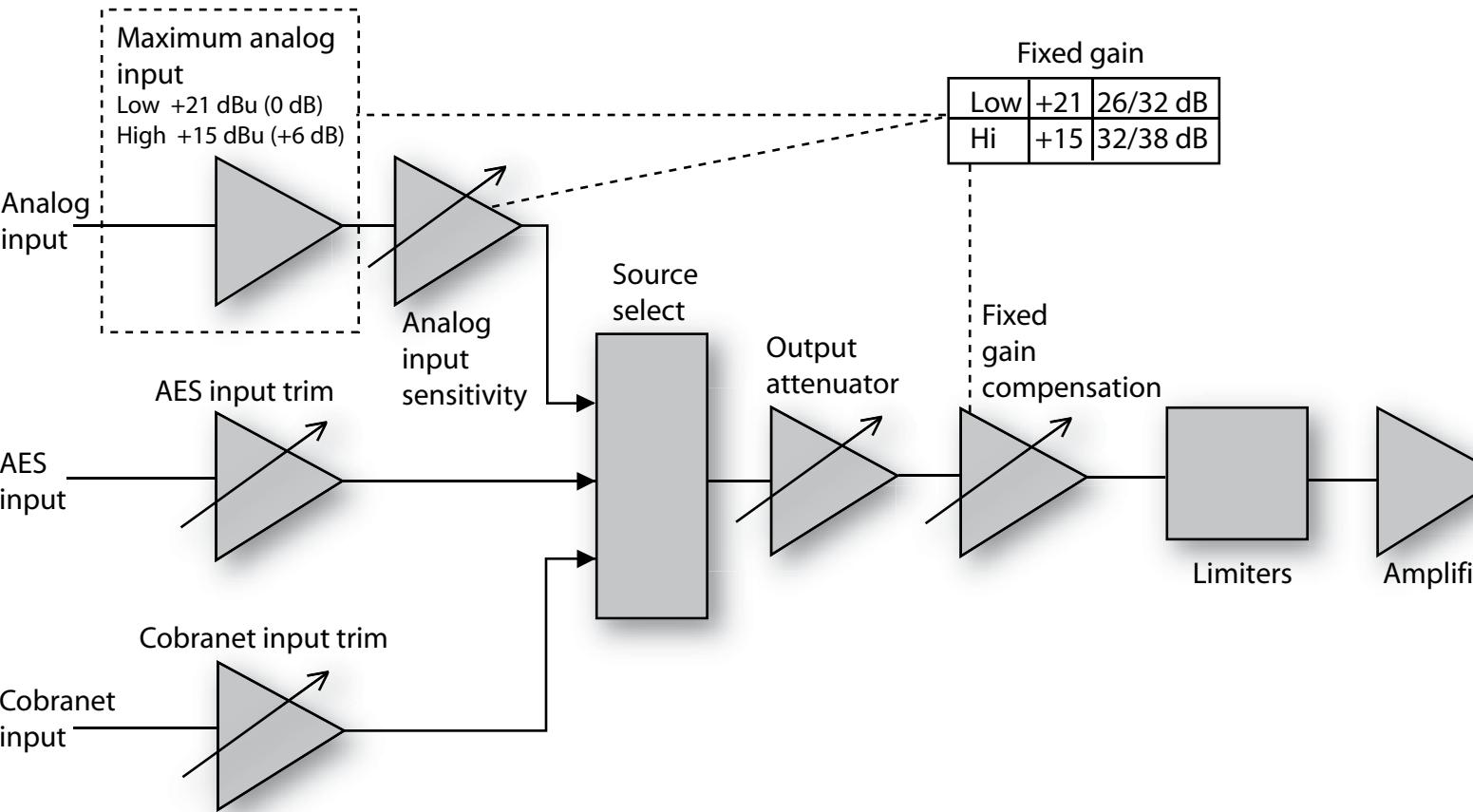
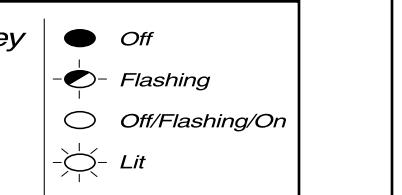
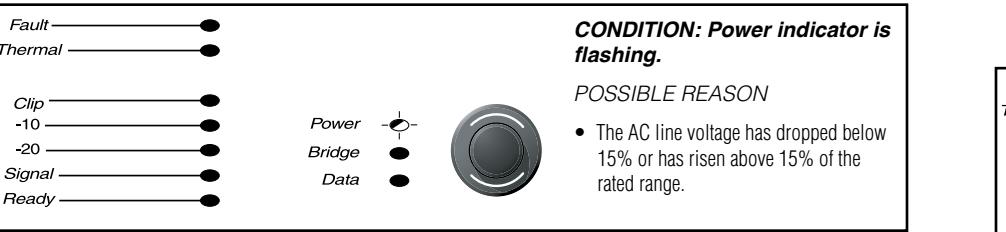
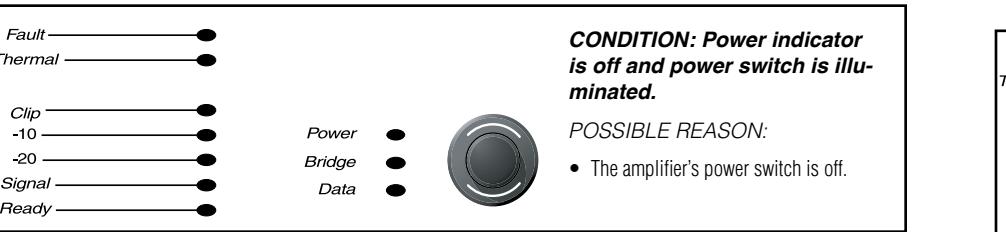
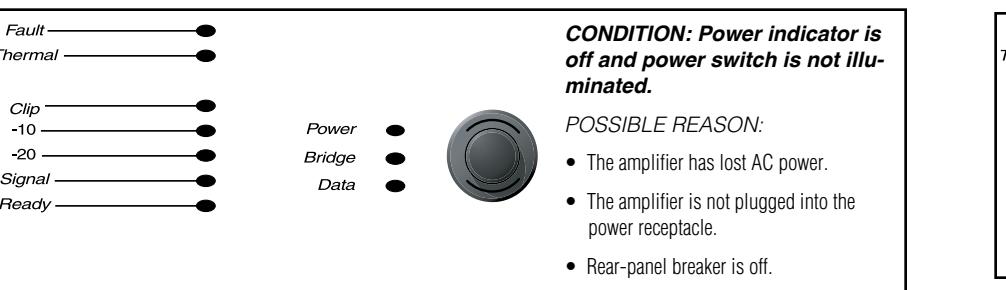
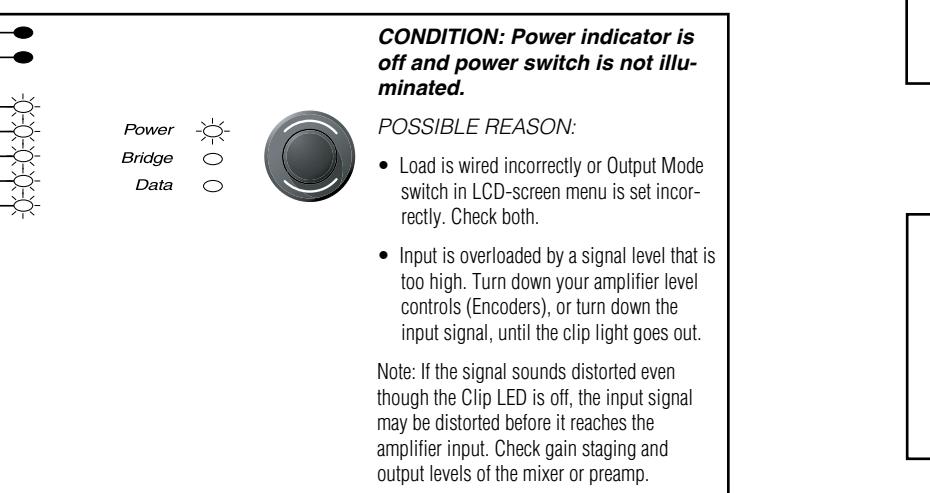
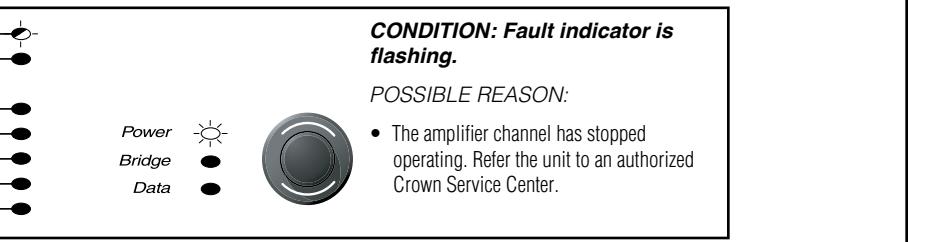
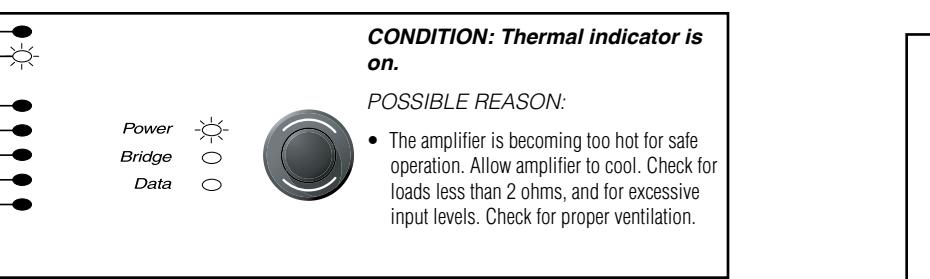


Figure 7.3 The Various Gain Settings in the I-Tech HD

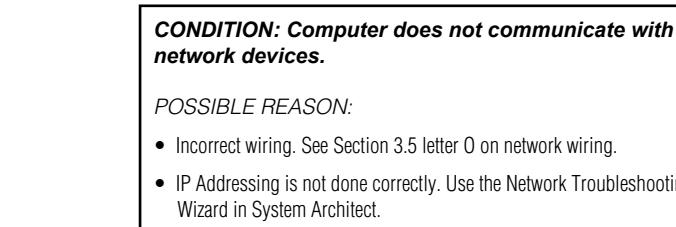
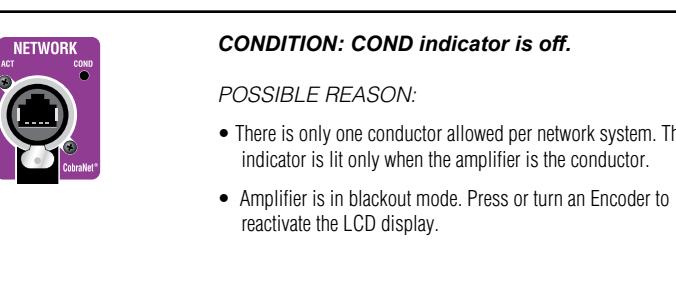
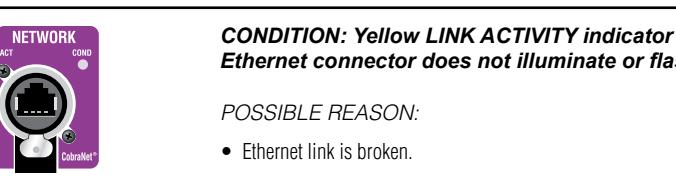
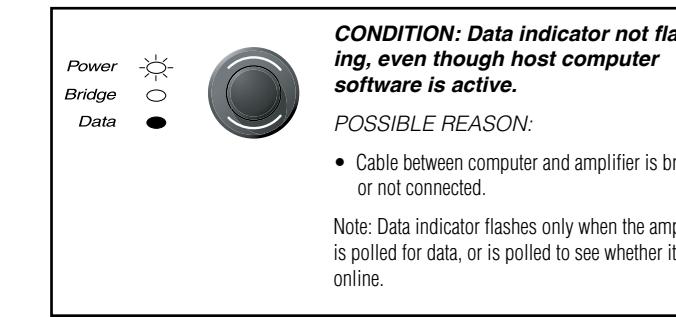
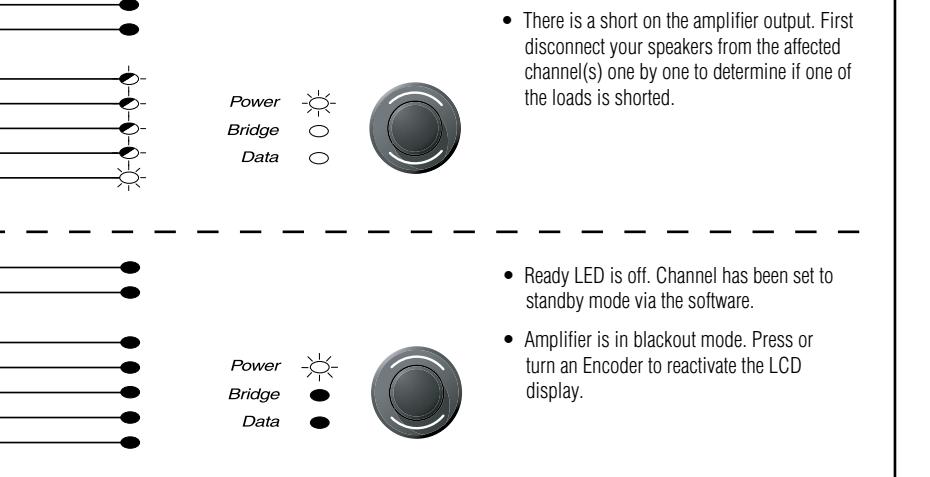
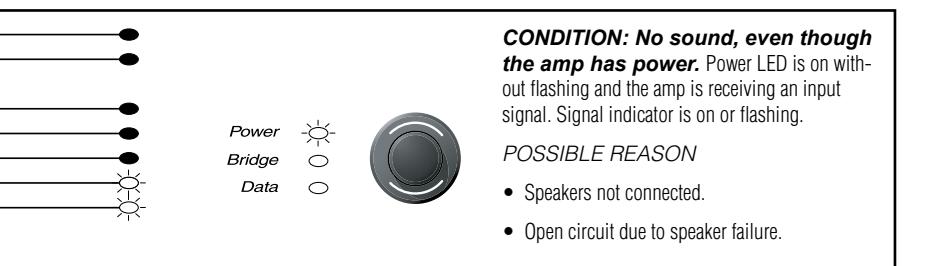
8 Troubleshooting



"Off/Flashing/On" means that the LED can be off, or flashing, or on.



8 Troubleshooting



9 Specifications

Minimum Guaranteed Power		
I-T4x3500HD		
20 Hz - 20 kHz with 0.35% THD, all channels driven	20 Hz - 20 kHz	1 kHz 20ms Burst
2-ohm	2,100W	3,500W
2.7ohm	2,200W	3,800W
4-ohm	2,400W	4,000W
8-ohm	1,900W	1,900W
4-ohm Bridge	4,200W	7,000W
8-ohm Bridge	4,800W	6,000W
70Vrms Direct	2,100W	
100Vrms Direct	2,300W	
Performance		
I-T4x3500HD		
Input Sensitivity (volts RMS) for rated output	Adjustable in 0.1V steps from 1.28V to 8V	
Voltage Gain for full rated power at 8 ohms	37.1 dB to 22.2 dB	
Frequency Response (20Hz to 20kHz at 1 watt, 20Hz - 20 kHz)	± 0.25 dB	
Signal to Noise Ratio below rated full-bandwidth power, A-weighted	> -108 dB	
Total Harmonic Distortion (THD) at full rated power at 1 kHz	< 0.35%	
Intermodulation Distortion (IMD) 60 Hz and 7 kHz at 4:1, from full rated output to -35 dB	< 0.06%	
Maximum Input Level	+15 dBu or +21 dBu, depending on setting of maximum input level	
Latency	1.16 mS at 48 kHz, 1.13 mS at 96 kHz	
A/D, D/A Converters	24-bit 96 kHz Cirrus Logic	
Digital Input	AES/EBU, 24-bit, 32-96 kHz. Onboard sample rate converter.	
Network	Onboard HiQnet and TCP/IQ, compatible with standard 100Mb Ethernet hardware	
DSP	24-bit conversion with 32-bit, floating-point DSP processing	
Attenuators	Speed-sensitive rotary encoder, 0.5 dB steps, range 0 to -100 dB.	

9 Specifications

Performance		I-Tech 4x3500HD
Damping Factor (8Ω): 20Hz to 100Hz		>5000
Crosstalk (below rated power, 20Hz to 1kHz)		> 80dB
Common Mode Rejection (CMR) (20 Hz to 1 kHz)		> 55dB, typically > 70dB
DC Output Offset (Shorted input)		< ± 3 mV
Input Impedance (nominal)		20 kilohms balanced, 10 kilohms unbalanced
Load Impedance (Note: Safe with all types of loads)		
Stereo		2-4-8-16Ω, 70V/100V
Bridge Mono		2-4-8Ω, 70V//100V
Required AC Mains		Universal AC input, 100-240VAC ±10%, 50/60Hz. Max. AC mains voltage.
AC Line Connector		USA, UK, European, Australia, India, Brazil, China (USA ships with Hubbell L6-30)
Construction		I-Tech 4x3500HD
Ventilation		Flow-through ventilation from front to back
Cooling		Dual-zone, microprocessor controlled, continuously variable speed fans
Dimensions		EIA Standard 19-inch rack mount width (EIA RS-310B), 3.5-inch (8.9-cm) height, 16.95-inch (43.1-cm) depth behind front mounting surface
Weight		
Net		29 pounds (13.1 kg)
Shipping		37 pounds (16.8 kg)

9 Specifications

Charts

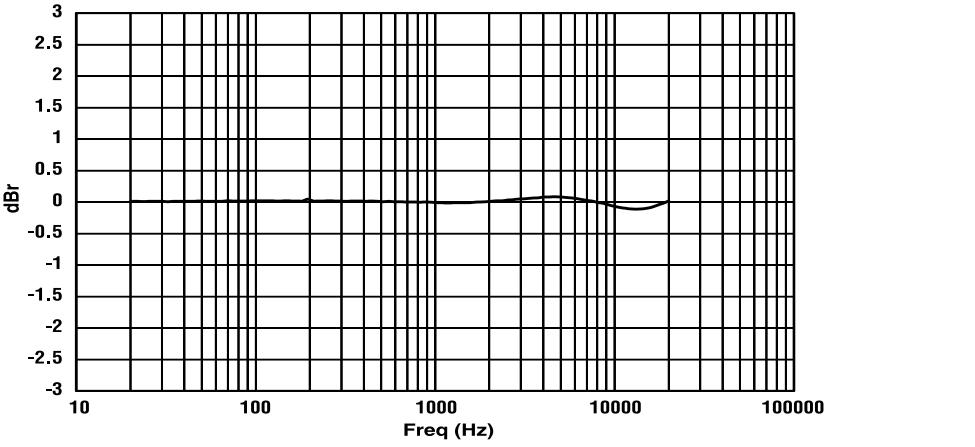


Figure 9.1 Typical Frequency Response (1W)

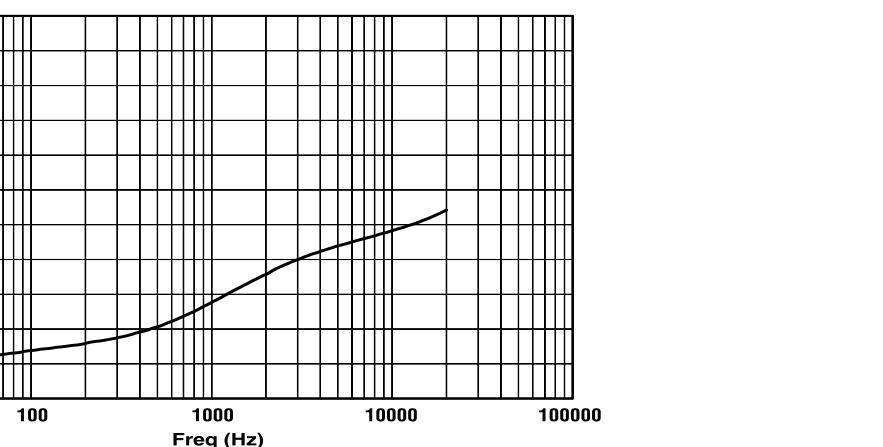


Figure 9.2 Typical Crosstalk vs. Frequency

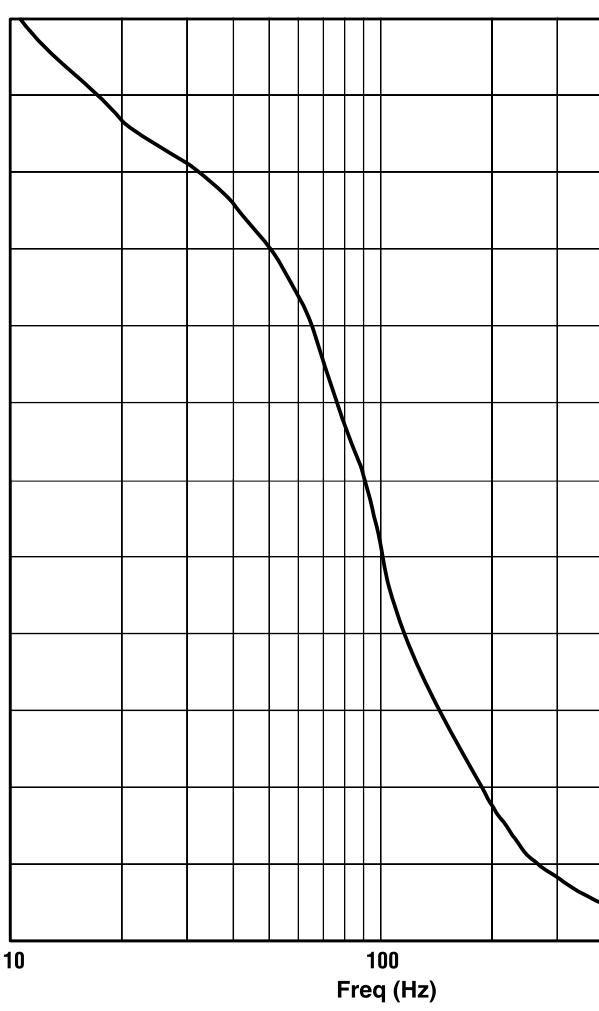


Figure 9.3 Typical Damping Factor vs. Frequency

9 Specifications

Operation Manual

10 AC Power Draw and Thermal Dissipation

I-Tech 4x3500HD AC Current Draw and Thermal Dissipation:

Pink noise 12dB crest factor, bandwidth limited 22Hz to 22kHz.

Typical line impedance used.

Data based on all channels driven.

I-T4x3500HD													
Condition	Load	120 VAC / 60 Hz			208 VAC / 60 Hz			240 VAC / 60 Hz					
		Line Current Amps	Power Dissipated as Heat		Line Current Amps	Power Dissipated as Heat		Line Current Amps	Power Dissipated as Heat		Watts	BTU	kcal/hr
			Watts	BTU		Watts	BTU		Watts	BTU			
Idle (sleep mode)		1.0	75	256	65	1.3	88	300	76	1.4	100	341	86
Idle (awake)		2.1	335	1143	288	2.1	328	1119	282	2.1	335	1143	288
1/8th Power Pink Noise Typical of program material just at clip.	8 Ohms/Ch.	13.5	559	1907	481	7.9	591	2017	509	7.6	620	2115	533
	4 Ohms/Ch.	16.0	787	2686	677	8.8	625	2133	538	7.2	766	2615	659
	2 Ohms/Ch.	17.0	886	3023	762	9.1	777	2652	669	8.8	979	3343	843
1/3rd Power Pink Noise Typical of program material at extreme clip.	8 Ohms/Ch.	32.2	962	3285	828	17.5	1180	4027	1015	16.5	1027	3506	884
	4 Ohms/Ch.	33.0	1084	3699	933	19.1	1092	3729	940	17.5	1092	3729	940
	2 Ohms/Ch.	36.8	1521	5191	1309	19.9	1228	4191	1057	19.6	1506	5140	1296

11 Advanced Features

11.1 Protection Systems

Your Crown amplifier provides extensive protection and diagnostic capabilities, including thermal level control, fault indicators, high-pass filtering, DC protect, AC under/over voltage protection, inrush limiting, and variable-speed fans with tachometer feedback. Microprocessor monitors fans, and signals an error via System Architect if fans are not operating.

11.1.1 Thermal Level Control (TLC)

If the amplifier becomes too hot for safe operation, the TLC will engage the input compressor and the clip LED will illuminate. By compressing the input, the amplifier will not generate as much heat and will have a chance to cool down. The degree of compression is proportional to the amount of overheating. This feature allows the show to go on, rather than having the amplifier shut down, and only occurs in extreme situations.

11.1.2 Circuit Breaker

If the current draw of the amplifier exceeds safe limits, this breaker automatically disconnects the power supply from the AC mains.

11.2 Global Switching Power Supply with PFC

Thanks to its global power supply, the I-Tech amplifier works anywhere in the world. There's no need to reset an AC mains voltage switch, and no need to order a special model. The amp will work and meet all specs on 100V - 240VAC, 50/60 Hz.

Crown's Switching Power Supply minimizes the amplifier's weight.

Typical non-switching power supplies require large, heavy transformers in order to produce the required power at the output stage. These transformers must be large to operate at 50 to 60 Hz (standard AC supplied by the power company).

By contrast, switching power supplies can operate with a much smaller (and lighter) transformer because they first convert the AC up to a much higher frequency, thereby reducing waste.

Power Factor Correction (PFC) controls how your amplifier draws current from the AC mains. Instead of drawing high-magnitude current spikes that reduce the capacity of your power

distro (and couple noise into other system components), PFC draws a smooth and quiet current waveform that is in phase with the mains voltage waveform. PFC allows you to reduce the size and weight of your power distribution and improve the performance of your signal processors.

11.3 6th Generation Class I Circuitry

Class I (BCA, Balanced Current Amplifier) Interleave technology allows high levels of power output without all the associated heat found in a conventional amplifier. This highly efficient design also provides accurate, uncolored sound with very low distortion and a high damping factor for tight, clean bass response. And it can deliver more power to reactive loads than conventional dissipative designs (comfortably driving the latest high-technology transducers). Class I also reduces current draw of the AC mains, providing energy conservation.

11.4 Color-Coded Rear Overlay

The labels on the rear panel are color coded to group similar functions under common colors.

12.2 A Closer Look at CobraNet

Licensed by Cirrus, CobraNet is a protocol, firmware and hardware that lets you transmit digital audio over a 100Base-T Fast-Ethernet network.

This section provides an overview of CobraNet. Cirrus Logic has provided a number of technical papers that describe CobraNet in detail. These papers are available at: <http://www.cobranetinfo.com>.

12.2.1 Fast Ethernet

(100Base-T)

Fast Ethernet (100Base-T) runs at 100Mbps (Megabits per second). It uses a hub or network switch to interconnect all network nodes (components). This forms a star network (see Figure 9.1). In a star network, all PCs are connected to a centralized hub or switch. Messages are sent from the originating PC to the hub, which then transmits the message either to the whole network or, if a switched hub (network switch) is used, to the destination device only.

In larger Fast Ethernet networks, additional hubs, concentrators, and other network hardware are used to form a larger network, as shown in Figure 12.4.

Today, commonly available networking cards are 10/100Base-T capable, which allows them to be used on either 10BaseT or 100Base-T networks.

The maximum length of cable for CobraNet is the same as for Ethernet: 100 meters over CAT-5 copper cable, 2 kilometers over multimode fiber. Proprietary "Fast Ethernet via single mode fiber" networks can reach even further.

12.2.2 Audio Specs

Sample rate: CobraNet can handle sample rates of 48 kHz or 96 kHz. I-Tech HD is switchable between 48 kHz and 96 kHz.

Bit depth: Set each channel to transmit 16-, 20- or 24-bit audio data as desired.

Latency: The transmitter and receiver buffers required for reliable transmission are set to 256 samples as a default. This gives a network transmission latency of 5.333 mS (256 samples at 1/48 kHz per sample). I-Tech HD latency is switchable between 1.33, 2.66, and 5.33 mS.

12.2.3 Bundles and Audio Channels

You assign all audio channels to groups called Bundles for transmission over the CobraNet network. Bundles were formerly called Network Channels.

A bundle can carry up to 8 audio channels. Up to 8 bundles can be sent in a 100 Mbps line, depending on the signal's bit depth and sampling rate. Only 4 bundles are available when 96 kHz sampling rate is in use.

The usual assignment is 8 channels at 20 bits. You can use fewer channels per bundle, but maximum size bundles are suggested for the most efficient use of network bandwidth. If 24-bit data is desired, then only 7 audio channels can be loaded into a single Bundle. With a 96 kHz sample rate, only 4 channels of audio are available.

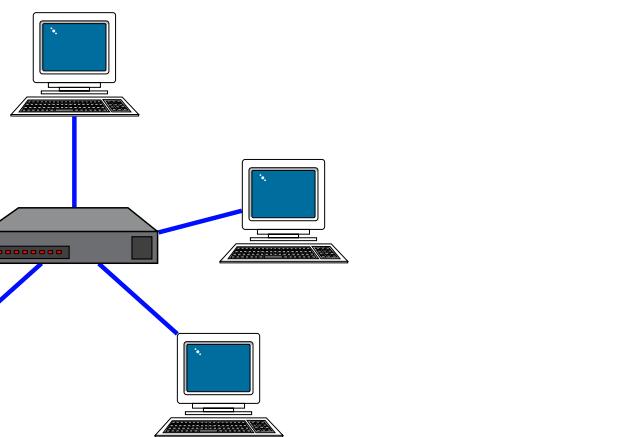


Figure 12.3 Star Topology

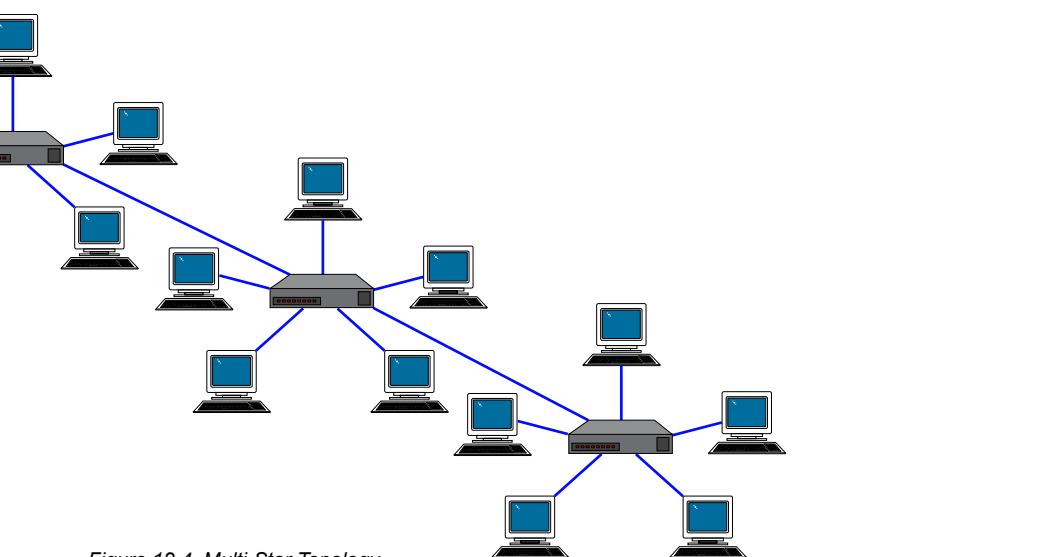


Figure 12.4 Multi-Star Topology

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13 LevelMAX™ Limiter Suite

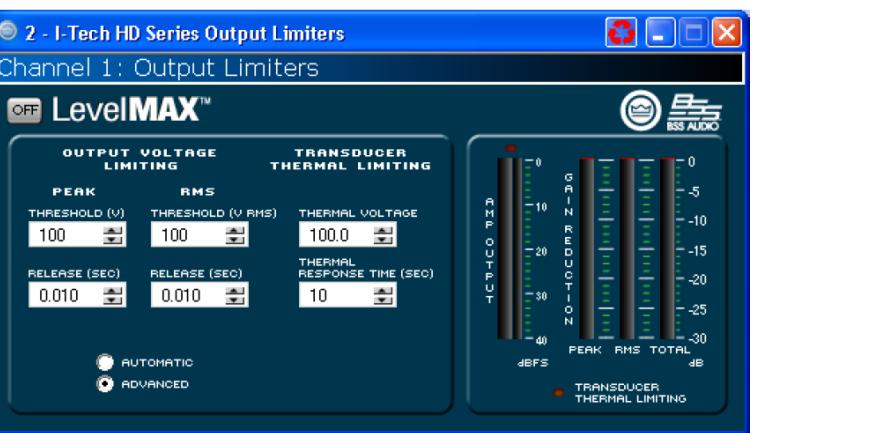
The i-Tech HD LevelMAX™ Limiter Suite features three types of limiting:

- Peak
- RMS
- Transducer Thermal

and three modes of operation:

- Auto
- Advanced
- Manual (OEM Factory Locked)

The standard i-Tech HD LevelMAX limiter panel (Advanced Mode) is shown below:



LevelMAX has been designed to limit accurately to the specified voltage thresholds. The addition of a thermal limiter allows thresholds for the faster-acting RMS and peak limiters to be set higher, providing more output and maintaining dynamic range before the onset of limiting, while protecting the long term integrity of the transducers.

In Auto or Advanced modes, turning on the LevelMAX Limiter Suite will enable all three limiters: Peak, RMS and Thermal.

The three LevelMAX limiter types and operating modes are outlined in the following sections.

14 LevelMAX Limiters

14.1 LevelMAX Peak Limiter

LevelMAX Peak limiting instantaneously limits the output voltage to not exceed the defined peak threshold voltage.

14.2 Peak Limiter Threshold (V_{pk})

In Auto mode, the peak limiter threshold value is automatically determined from the defined RMS threshold and the low pass frequency defined in the XOVER block. For a low pass frequency less than 100 Hz, the RMS to peak threshold ratio is 6 dB. From 100 Hz to 400 Hz, the RMS to peak ratio increases from 6 dB to 9 dB. Above 400 Hz, the ratio remains at 9 dB.

For Advanced and Manual modes, the peak limiter threshold can be set by the user.

14.3 Peak Limiter Attack (sec)

The peak limiter attack time is instantaneous and cannot be modified. The output voltage is limited to the value defined by the peak limiter threshold (see 14.1).

14.4 Peak Limiter Release (sec)

In Auto mode, the peak limiter release time is determined by the high pass frequency defined in the XOVER block.

For Advanced and Manual modes, the peak limiter release time can be set by the user.

14.5 Peak Limiter Look-Ahead

Peak Limiter Look-Ahead inserts a delay into the peak limiter signal path in order to 'look' at the input signal and anticipate when the peak threshold will be exceeded. If it detects that the threshold will be exceeded, the signal is gradually reduced prior to reaching the threshold, while still limiting to the correct peak voltage. This approach eliminates a sharp signal discontinuity which, in turn, could create audible distortion.

The peak limiter look-ahead value is 0.5 ms. This value was determined to be subjectively acceptable for most frequency ranges while not adversely compromising overall latency. Overall latency remains

constant regardless of whether the LevelMAX Limiter suite is enabled (ON) or bypassed (OFF).

14.6 LevelMAX RMS Limiter

LevelMAX RMS limiting accurately limits the output signal to the specified RMS threshold voltage.

14.7 RMS Threshold (VRMS)

The RMS Threshold should be set by the user to correspond to the short term (2 hour) power handling of the transducer or system.

Refer to the appropriate JBL Specification Sheet, Technical Manual or Product Manual for two hour power ratings. To calculate the RMS Threshold in VRMS, use the following equation:

$$\text{RMS Threshold (Vrms)} = \sqrt{\text{AES Power (Wrms, 2 hour)} \times \text{Minimum Impedance (ohms)}}$$

where AES Power (Wrms, 2 hour) is the free air component power rating.

If minimum impedance data is unavailable, use nominal impedance. If 2 hour AES power ratings are unavailable, use 2 hour IEC power ratings. For most applications, it is recommended to operate LevelMAX limiters in Auto mode so that peak thresholds are automatically calculated based on the RMS threshold and selected LP frequency (see 14.1).

14.8 LevelMAX Transducer Thermal Limiter

LevelMAX Transducer Thermal limiting is designed to protect transducers from long term thermal damage by gradually adjusting the RMS threshold until the target long term thermal threshold voltage has been reached. The thermal limiter functions only when the RMS limiter is enabled and when the thermal voltage threshold is lower than the RMS threshold. On the main limiter panel, the thermal protection LED turns on once the thermal limiter is 0.75 dB into limit.

14.9 Thermal Voltage (V_{RMS})

The thermal voltage threshold should be set to correspond to the long term (100 hour) power handling of the transducer or system.

Refer to the appropriate JBL Specification Sheet, Technical Manual or Product Manual for 100 hour power ratings. To calculate the Thermal Voltage Threshold in VRMS, use the following equation:

$$\text{Thermal Threshold (Vrms)} = \sqrt{\text{Power (Wrms, 100 hour)} \times \text{Nominal Impedance (ohms)}}$$

where Power (Wrms, 100 hour) is the 100 hour power rating.

If 100 hour power handling data is unavailable, set 100 hour power rating 0.8 x 2 hour power rating.

14.10 Thermal Response Time (s)

The thermal response time is the amount of time the average RMS signal is above the thermal threshold voltage before limiting begins. Once the threshold has been exceeded for the amount of time defined by the thermal response time, the RMS threshold is lowered until the target thermal voltage has been reached. The thermal response time is not the time it takes to reach the target voltage, although the longer the thermal response time, the longer it takes to reach the target voltage. It is also dependent on the amount of gain reduction and the ratio between RMS and thermal thresholds.

Thermal response time is transducer-dependent and should be based on the initial temperature rise of the transducer voice coil to protect it from thermal overload.

If detailed transducer data is unavailable, the Thermal Response Time should be left at the default value of 10 sec.

15 Appendix B: Table of Parameters Modified by Each Mode with LevelMAX Limiter Suite Enabled

Parameter	Auto	Advanced	Visible in Main Panel
Peak Enable	✓	✓	
Peak Threshold	✓		✓
Peak Attack	✓	✓	
Peak Release	✓		✓
RMS Enable	✓	✓	
RMS Threshold			✓
RMS Attack	✓	✓	
RMS Release	✓		✓
RMS Detector Time	✓	✓	
Thermal Enable	✓	✓	
Thermal Threshold			✓
Thermal Attack			✓
Thermal Slope (Kp & Ki)	✓	✓	
Look-Ahead	✓	✓	

SV Value Range

Parameter	Auto	Increment
Peak Threshold (See Below)	1-400V	1V
Peak Release	0.1ms-10s	1ms
RMS Threshold	1-200V	1V
RMS Attack	0-10s	n/a
RMS Release	0-10s	1ms
RMS Detector Time	1ms-1s	n/a
Thermal Threshold	1-200V	0.1V
Thermal Attack	1-600V	1s
Thermal Slope (Kp & Ki)	1-10	n/a
Look-Ahead	0-10ms	n/a

Auto Peak max for each model (online)

Parameter	Auto
I-Tech HD 5000	201.2
I-Tech HD 9000	217.6
I-Tech HD 12000	233.4
I-Tech 4x3500 HD	217.6

16 Application of FIR Filters to Loudspeaker Crossovers

16.1 FIR Overview

The powerful DSP processing in the latest Crown I-Tech HD Series amplifiers allow sophisticated Finite Impulse Response filters as well. FIR filters offer a number of very strong advantages when used to implement DSP electronic loudspeaker crossovers.

Although high-order IIR filters, which are based on conventional analog filters, can be designed to provide high stop-band roll-off rates, the resultant filter phase response is highly nonlinear. This significantly complicates crossover design and implementation.

As compared to IIR filters, FIR filters can be straightforwardly designed to allow extremely narrow crossover overlap between adjacent drivers with high stop-band attenuation and very-high roll-off rates. In-phase and linear-phase crossovers can easily be designed. Minimizing overlap dramatically reduces polar lobing and vastly improves off-axis response through the crossover region.

This white paper briefly describes the characteristics and pros/cons of FIR and IIR filters, discusses the desirable attributes of FIR filters, and closes with a set of example measurements on a two-way loudspeaker system to illustrate the practical application of IIR and FIR crossover filters. In addition, audio demonstration files are available on Crown's web site that demonstrates the effects described in this white paper (see comments at the end of this report).

16.2 What are IIR Filters and FIR Filters?

A filter modifies certain characteristics of a signal such as amplitude/phase frequency response and wave-shape in a desired manner. This can be done either in a purely analog manner with a real piece of hardware such as a non-computer based equalizer or filter set, or with a computer-based instrument running mathematical algorithms using digital signal processing techniques. These algorithms can be either implemented in hardware and/or software. The term "digital filter" refers to the specific hardware or software routine that performs the filtering algorithm.

Digital filters may be further classified into two general types: Infinite Impulse Response (IIR) and Finite Impulse Response (FIR). Fundamentally, IIR filters include feedback (recursion) in their implementation while FIR filters do not. The feedback in an IIR filter means that the impulse response of the filter theoretically goes on forever, thus the "Infinite" term in the name. The impulse response of a FIR filter, on the other hand, is finite because there is no feedback. Simplified structures for both these types of filters are shown in Fig. 16.1.

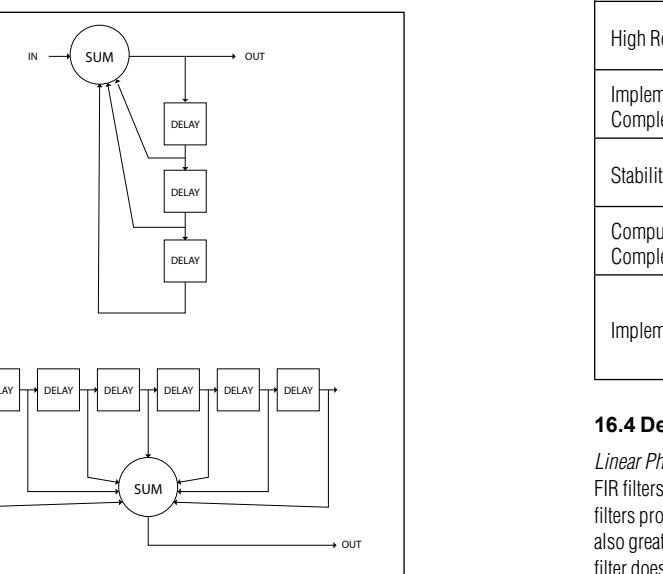


Fig. 16.1 Simplified filter structures of an IIR (top) and FIR (bottom) filter

The IIR structure on the left feeds back weighted the delayed versions of the output signal back to the input and thus sets up a form of recirculation which effectively goes on for forever once a signal is applied to the input. The FIR structure on the right feeds weighted and delayed signals to the output only and thus can't recirculate the signals. Effectively there is only a single path through the FIR filter.

16 Application of FIR Filters to Loudspeaker Crossovers

16.3 Pros and Cons of IIR and FIR Filters

The following table lists several characteristics of the two types of filters and their pros and cons.

Characteristic	IIR Filter	FIR Filter
Linear Phase Response	Not Possible	Possible
High Rolloff Rates	Yes, but with phase distortion	Yes, without phase distortion
Implement Complex Filters	No	Yes
Stability	Conditionally	Naturally Stable
Computational Complexity	Few CPU cycles	Massive CPU cycles
Implementation	Common Practice	Difficult to achieve with limited DSP

16.4 Desirable Attributes of FIR Filters

Linear Phase

FIR filters can be designed to have exact linear phase. Linear-phase filters provide minimal modification to the wave shape of a signal and also greatly simplify crossover design and implementation because the filter does not change the phase of the drivers being crossed over.

A linear phase crossover does not mean that the overall response of the crossover including driver response is linear phase. This can be true only if the individual drivers themselves are linear phase or can be equalized to be linear phase.

Electronic crossovers using conventional non-linear phase filters such as IIR filters can artificially increase the crest factor of a signal and thus decrease the headroom in the transmission channel.

16.5 High Rolloff and Steep Slopes

FIR filters can be designed to have extremely high stop-band roll-offs and exceptionally steep slopes which greatly minimizes crossover driver overlap. In a conventional analog or analog-based IIR crossover, driver overlap can extend over two or three octaves. FIR crossovers dramatically restrict the operating overlap bandwidth of the crossover which considerably reduces the range over which both upper and lower range drivers are radiating in the same frequency range. Very narrow overlaps of one-third octave or less can be implemented with FIR filters.

In addition, the extremely steep slopes of FIR filters offer greater driver protection and reduced distortion. Beyond the driver's linear frequency range, energy is attenuated so rapidly that most non-linearity's cease to be a problem. The driver does not need to be as well behaved outside its frequency range. Power handling capability of HF drivers is much improved. The narrower crossover region also lessens the need for precise driver time alignment since the overlap region is so small.

16.6 Stop-band Attenuation

Associated with the very high stop-band rolloff of an FIR filter, is the associated extremely high stop-band attenuation. This minimizes interaction between adjacent drivers such as a low-frequency woofer signal bleeding into a tweeter and thus causing intermodulation distortion. In a home theater setup, high stop-band attenuation of the subwoofer minimizes subjective localization of the woofers due to high-frequency bleed through.

16.7 Polar Lobing Error

Polar lobing is a potential problem when spatially separated non-coincident drivers are crossed over. Through the crossover region both drivers are radiating simultaneously. This may cause a narrowing of the coverage pattern and the creation of a directional lobe. It is very desirable that this lobe face straight ahead, and not directionally wander with frequency. If it does, lobing error occurs.

Polar lobing error is minimized when the low- and high-pass sections of the crossover are in-phase with each other throughout the crossover are in-phase with each other throughout the crossover region. This is an attribute of the "Linkwitz-Riley" (LR) type of crossover responses. Polar lobing can also be minimized by reducing the crossover overlap with zero-phase sharp-cutoff filters such as provided by FIR filters.

16.8 Crown's Implementation of FIR Filters

Crown's FIR filter implementation uses state-of-the-art digital signal processing techniques which are highly optimized for the DSP engine in the I-Tech HD series of amplifiers.

16.9 FFT Convolution

The resource requirements of a high-performance FIR filter in terms of CPU cycles per sample, datapath bandwidth and memory footprint can exceed those of an IIR solution by several orders of magnitude. Key to an efficient FIR implementation is the use of Fast Fourier Transform (FFT) techniques to accelerate the FIR convolution process, which is usually thought of as a time domain operation. Time domain convolution is something to be avoided, because it is extremely expensive in the computational sense. Fortunately, signal processing theory tells us that multiplication in the frequency domain is equivalent of convolution in the time domain. This is important because multiplication is very efficient in comparison to convolution. Of course, an efficient and speedy means of moving between the time and frequency domains is also required. This is where the FFT comes in. Using the FFT to transform back and forth between the time and frequency domains so as to replace convolution with multiplication is referred to as FFT Convolution. Figure 16.2 shows a block diagram of the FFT convolution process.

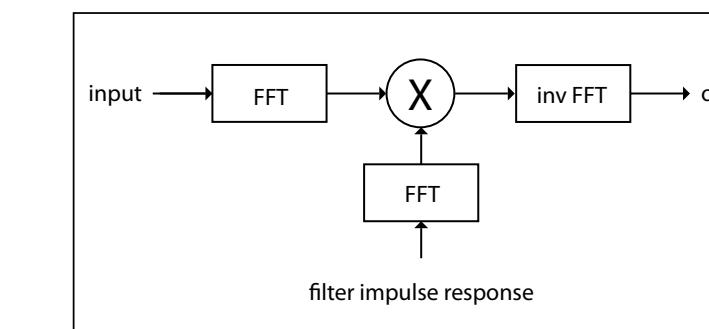


Fig. 16.2 FFT convolution block diagram. here the input and filter impulse responses are both individually FFT'd and multiplied and then inverse FFT'd to generate the output.

16 Application of FIR Filters to Loudspeaker Crossovers

16.10 Filter Design

IIR filters are typically based on an equivalent analog circuit. Because of this, the IIR design process is a straightforward matter of converting an analog filter frequency response specification into digital coefficients. In contrast, FIR frequency response is usually specified in terms of an ideal, though physically unrealizable filter, e.g. a brick wall lowpass. The FIR filter design is the process of generating coefficients that approximate the response of the ideal filter. The I-Tech HD FIR filters are designed using a proprietary iterative algorithm using state-of-the-art signal processing techniques. It yields linear response and predictable low-noise operation over the entire audio frequency range.

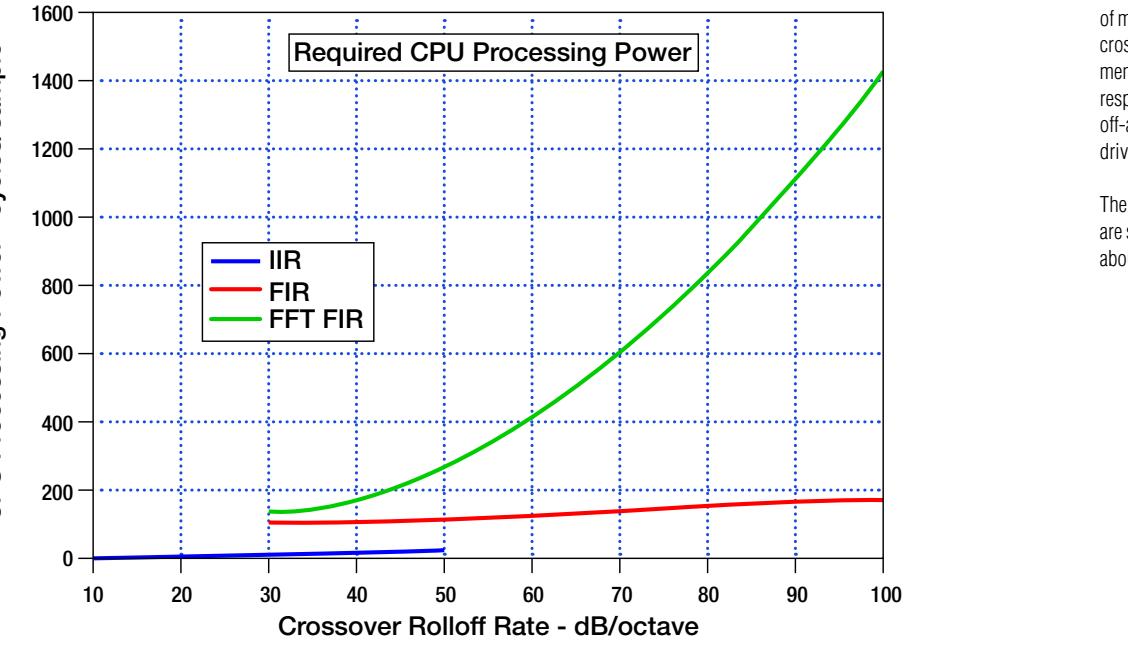


Fig. 16.3 FFT convolution block diagram. here the input and filter impulse responses are both individually FFT'd and multiplied and then inverse FFT'd to generate the output.

16.11 Low Latency 96 kHz Studio Quality Filters

Because of the extreme computational load imposed by high rolloff FIR crossover filters, a potent DSP and highly optimized FFT Convolution algorithm are essential to low latency operation at 96 kHz. In-system performance is equally dependent on the filter design process itself. I-Tech HD's platform-optimized FFT Convolution algorithm and state-of-the-art filter design methodology combine to deliver low latency 96 kHz studio quality filters.

Real World Benefits

16.12 Measurements of Two-way Loudspeaker System

The practical advantages of using an FIR-based crossover can be illustrated by comparing results with an IIR crossover when a two-way loudspeaker is setup and measured. For measurement purposes, a small two-way loudspeaker was driven by a Crown I-Tech 12000 HD power amplifier controlled by Crown's HiQnet™ System Architect™ software. The use of a small system in this application ensures that the frequency response and angular coverage of the individual drivers has sufficient overlap through the crossover region to properly illustrate frequency response and crossover effects.

The following two sets of measurements show the results of using a conventional IIR filter set up as a 4th-order Linkwitz-Riley crossover and a FIR filter set up as a linear-phase high-rolloff crossover. Each set of measurements first show the magnitude and phase response of the crossover alone and then follows the frequency response measurements of the system itself showing individual driver responses, on-axis responses with the crossover in and out of polarity, and a single off-axis frequency response where the response exhibits a null due to driver spacing.

The small two-way system measured here has a woofer and tweeter that are separated by about 5.5" which creates an off-axis polar null at about $\pm 25^\circ$ at the 3 kHz crossover frequency.

Note that the individual driver responses were first equalized using the amplifier's parametric equalizer to be more or less flat over a significantly wide range above and below crossover. After the initial equalization, the chosen crossover was applied and adjusted. Slight delay was added to the tweeter channel to compensate for driver offset to insure that the high- and low- pass sections were in phase through crossover.

One unexpected benefit of the FIR crossover was the speed up of the adjustments to insure that the drivers were in-phase through crossover. This was because the linear/zero phase characteristic of the FIR filter did not change the phase of the individual drivers when the crossover was applied! The linear phase characteristic of the FIR-filter based crossover significantly reduces the setup time of the crossover as compared to an IIR implementation.

All measurements were accomplished using windowed freefield techniques in a non-anechoic room at 1m using a 25 ms window. This window size allows fairly-accurate measurements to be made down to about 100 Hz.

IIR Measurements

A 4th-order Linkwitz-Riley (LR) crossover with 24 dB/octave slopes at 3 kHz was chosen to demonstrate a typical well-designed conventional crossover. Although, in conventional terms, a relatively sharp 24 dB/octave crossover slope is considered quite sharp. The crossover region spans a relatively wide two octaves. This means that effectively both woofer and tweeter are radiating simultaneously between roughly 1.5 and 6 kHz.

The following two graphs in Fig. 16.4 show the theoretical frequency response magnitude (left) and phase (right) of a 3 kHz LR 4th-order crossover. In each graph, individual curves are shown for the low pass (blue) and high pass (red) portions of the crossover. Note that both low-pass and high-pass filters are 6 dB down and in-phase at the 3 kHz crossover and thus sum to unity. Note also that the phase of the low- and high- pass sections is identical but exhibit a 360° phase rotation through crossover. The jump in phase at 3 kHz is a byproduct of the wrapped phase display, which keeps the phase on a $\pm 180^\circ$ scale. The LR responses will sum to unity at all frequencies as expected, but will exhibit an all-pass non-linear phase characteristic.

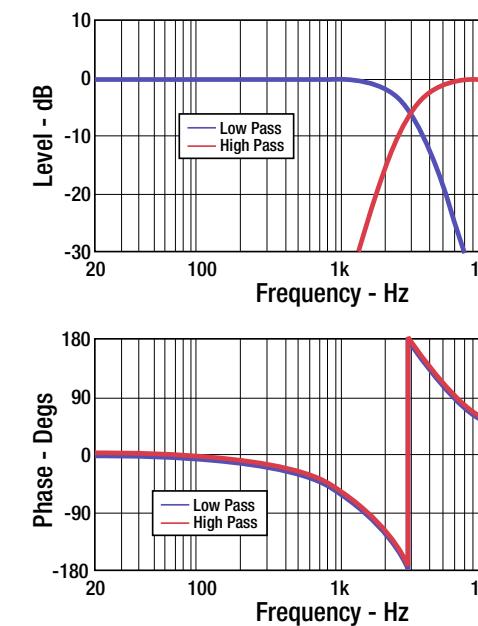


Fig. 16.4 Magnitude (left) and phase (right) of a 3 kHz Linkwitz-Riley 4th-order crossover filter.

The next two graphs in Fig. 16.5 show the measured individual driver responses of the two-way system driven by the LR crossover as produced by the Crown I-Tech HD amplifier and the resultant summed 1m on-axis response.

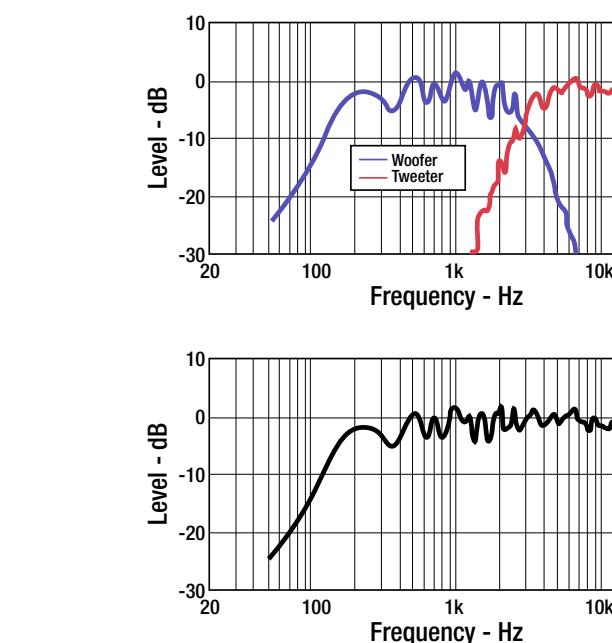


Fig. 16.5 Measured individual driver responses (left) and overall summed on-axis response (right) of a two-way system with a 4th-order 3 kHz LR crossover.

16 Application of FIR Filters to Loudspeaker Crossovers

The remaining two LR crossover graphs in Fig. 17.6 show the reversed-polarity on-axis response and the $\pm 25^\circ$ off-axis response. Note that when the tweeter leads are reversed, the response exhibits a deep null at the 3 kHz crossover. However, the null is also accompanied by a wide two-octave depression in response. The second graph of the pair shows the 25° off-axis null due to driver spacing. Not surprisingly, the response is similar to the on-axis response when the polarity is reversed.

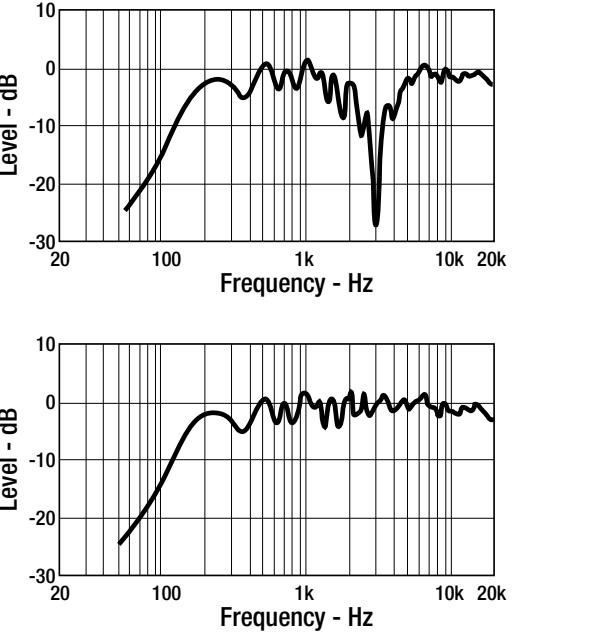


Fig. 16.6 Reversed-polarity on-axis response (left) and $\pm 25^\circ$ off-axis response of a two-way system with a 4th-order 3 kHz crossover.

16.13 FIR Measurements

The Crown I-Tech HD amplifier was used to create a linear-phase FIR crossover with a 3.67 ms latency. This latency allows crossovers to be created with filters whose corner frequencies can extend down 500 Hz. Selections of higher latencies allow sharper cutoff filters to be created with correspondingly lower corner frequencies. FIR filters with latencies from 0.92 ms up to 29.33 ms can be created with the amplifier's software with minimum corner frequencies ranging from 2 kHz down to 62.5 Hz. The 3.67 ms latency FIR filter provides a relatively sharp rolloff filter with reasonably short latency.

Figure 7 shows two graphs illustrating the measured frequency response magnitude (left) and phase (right) of the 3.67 ms latency FIR 3 kHz crossover generated by the HD amplifier. The right graph shows the measured phase of the crossover with the phase rotation due to the latency and speaker-to-microphone transit time subtracted out. It shows a linear zero-phase characteristic over most of the measured range.

Note the extremely fast filter roll-offs in the left graph which approach 150 dB/octave and the exceptionally-narrow one-third octave crossover span.

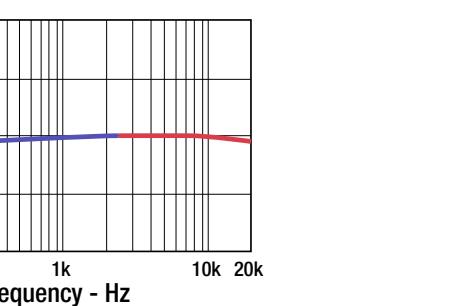
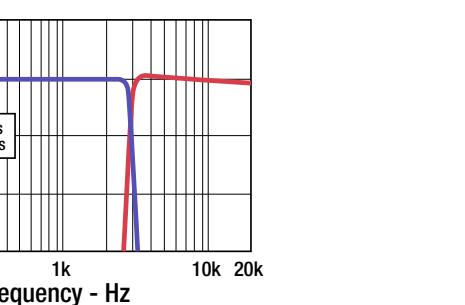


Fig. 16.7 Magnitude (left) and phase (right) of a 3.67 ms latency 3 kHz FIR crossover filter generated by an I-Tech HD amplifier.

Figure 16.8 shows the measured individual driver responses of the same two-way system driven by the FIR crossover as produced by the Crown I-Tech HD amplifier and the resultant summed 1m on-axis response. Note the extremely sharp filter roll-offs in the left magnitude graph. Note also that the right on-axis response is essentially identical to the previous LR on-axis response.

Figure 16.8 shows the measured individual driver responses of the same two-way system driven by the FIR crossover as produced by the Crown I-Tech HD amplifier and the resultant summed 1m on-axis response. Note the extremely sharp filter roll-offs in the left magnitude graph. Note also that the right on-axis response is essentially identical to the previous LR on-axis response. The final two graphs, shown in Fig. 17.9, illustrate the reversed-polarity on-axis response and the $\pm 25^\circ$ off-axis response for the FIR crossed-over system. Note the extremely narrow crossover regions due to the rapid filter rolloff of the FIR filter. Psycho acoustic tests reveal that these narrow high-Q dips in the response are much less audible than low-Q wider dips.

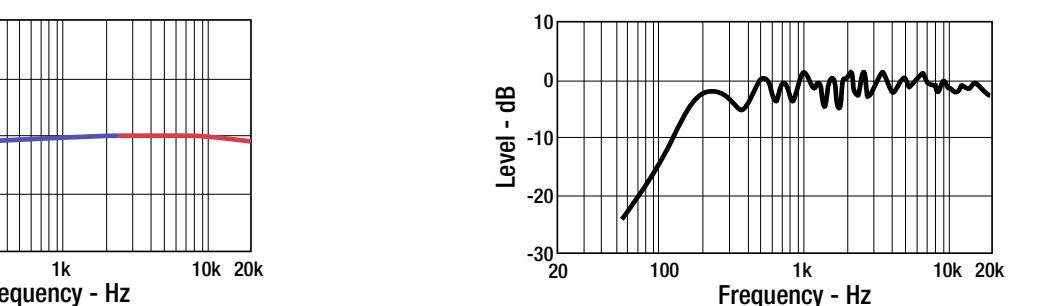
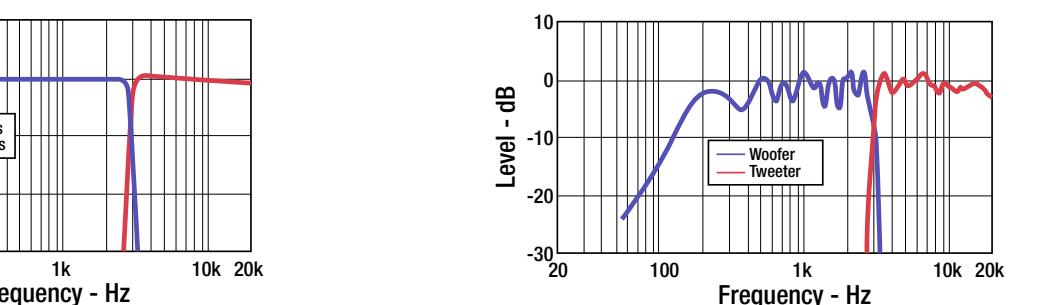


Fig. 16.8 Measured individual driver responses (left) and overall summed on-axis response (right) of the two-way system of Fig. 5 driven with 3.67 ms latency 3 kHz FIR crossover filter generated by an I-Tech HD amplifier.

Figure 16.8 shows the measured individual driver responses of the same two-way system driven by the FIR crossover as produced by the Crown I-Tech HD amplifier and the resultant summed 1m on-axis response. Note the extremely sharp filter roll-offs in the left magnitude graph. Note also that the right on-axis response is essentially identical to the previous LR on-axis response.

The final two graphs, shown in Fig. 17.9, illustrate the reversed-polarity on-axis response and the $\pm 25^\circ$ off-axis response for the FIR crossed-over system. Note the extremely narrow crossover regions due to the rapid filter rolloff of the FIR filter. Psycho acoustic tests reveal that these narrow high-Q dips in the response are much less audible than low-Q wider dips.

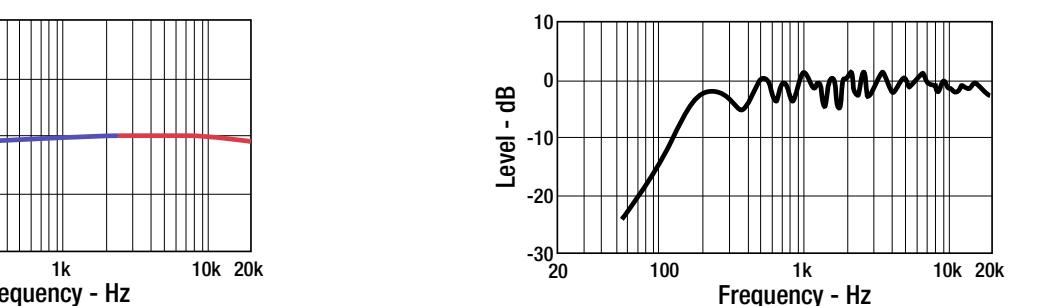
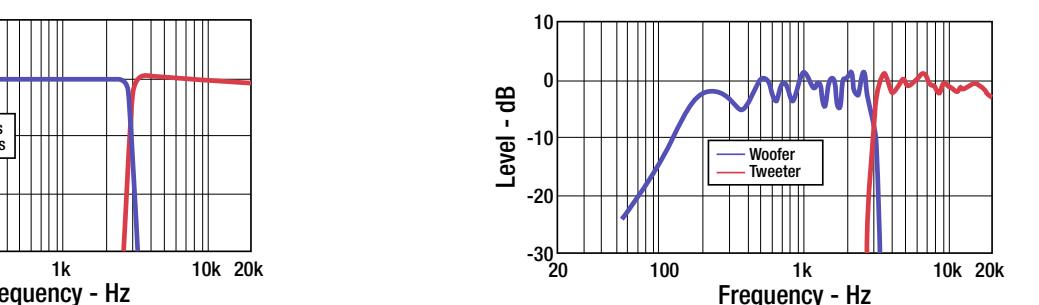


Fig. 16.9 Reversed-polarity on-axis response (left) and $\pm 25^\circ$ off-axis response of the two-way system of Fig. 5 driven with 3.67 ms latency 3 kHz FIR crossover filter generated by an I-Tech HD amplifier.

Figure 16.8 shows the measured individual driver responses of the same two-way system driven by the FIR crossover as produced by the Crown I-Tech HD amplifier and the resultant summed 1m on-axis response. Note the extremely sharp filter roll-offs in the left magnitude graph. Note also that the right on-axis response is essentially identical to the previous LR on-axis response.

The final two graphs, shown in Fig. 17.9, illustrate the reversed-polarity on-axis response and the $\pm 25^\circ$ off-axis response for the FIR crossed-over system. Note the extremely narrow crossover regions due to the rapid filter rolloff of the FIR filter. Psycho acoustic tests reveal that these narrow high-Q dips in the response are much less audible than low-Q wider dips.

17 Service

Crown amplifiers are quality units that rarely require servicing. Before returning your unit for service, please contact Crown Technical Support to verify the need for servicing.

This unit has very sophisticated circuitry which should only be serviced by a fully trained technician. This is one reason why each unit bears the following label:

CAUTION: To prevent electric shock, do not remove covers. No user serviceable parts inside. Refer servicing to a qualified technician.



Complete the Service Return Request form, in the back of this manual, when returning a Crown product to the factory or authorized service center. The form must be included with your product inside the box or in a packing slip envelope securely attached to the outside of the shipping carton. Do not send this form separately.

*Warranty is only valid within the United States of America. For information on Warranty outside of the U.S.A, please contact your local distributor.

17.1 International and Canada Service

Service may be obtained from an authorized service center. (Contact your local Crown/Amcor representative or our office for a list of authorized service centers.) To obtain service, simply present the bill of sale as proof of purchase along with the defective unit to an authorized service center. They will handle the necessary paperwork and repair.

Remember to transport your unit in the original factory pack.

17.2 US Service

Service may be obtained in one of two ways: from an authorized service center or from the factory. You may choose either. It is important that you have your copy of the bill of sale as your proof of purchase.

17.2.1 Service at a US Service Center

This method usually saves the most time and effort. Simply present your bill of sale along with the defective unit to an authorized service center to obtain service. They will handle the necessary paperwork and repair.

Remember to transport the unit in the original factory pack. A list of authorized service centers in your area can be obtained from Crown Factory Service, or online from www.crownaudio.com.

17.2.2 Factory Service

Crown accepts no responsibility for non-serviceable product that is sent to us for factory repair. It is the owner's responsibility to ensure that their product is serviceable prior to sending it to the factory. For more information, please contact us direct.

A Service Return Authorization (SRA) is required for product being sent to the factory for repair. An SRA can be completed online at www.crownaudio.com. If you do not have access to the web, please call Crown's Customer Service at 574.294.8200 or 800.342.6939 extension 8205.

For warranty service, we will pay for ground shipping both ways in the United States. Contact Crown Customer Service to obtain prepaid shipping labels prior to sending the unit. Or, if you prefer, you may prepay the cost of shipping, and Crown will reimburse you. Send copies of the shipping receipts to Crown to receive reimbursement. Your repaired unit will be returned via UPS ground. Please contact us if other arrangements are required.

17.2.3 Factory Service Shipping Instructions:

1. Service Return Authorization (SRA) is required for product being sent to the factory for service. Please complete the SRA by going to www.crownaudio.com. If you do not have access to our website, call 1.800.342.6939, extension 8205 and we'll create the SRA for you.
2. See packing instructions that follow.
3. Ship product to:
CROWN AUDIO FACTORY SERVICE
1718 W MISHAWKA RD.
ELKHART, IN 46517
4. Use a bold black marker and write the SRA number on three sides of the box.

5. Record the SRA number for future reference. The SRA number can be used to check the repair status.

17.2.4 Packing Instructions

Important: These instructions must be followed. If they are not followed, Crown Audio, Inc. assumes no responsibility for damaged goods and/or accessories that are sent with your unit.

1. Fill out and include the Service Return Request form in the back of this manual or at www.crownaudio.com.

2. Do not ship any accessories (manuals, cords, hardware, etc.) with your unit. These items are not needed to service your product. We will not be responsible for these items.

3. When shipping your Crown product, it is important that it has adequate protection. We recommend you use the original pack material when returning the product for repair. If you do not have the original box, please call Crown at 800.342.6939 or 574.294.8210 and order new pack material. See instructions for "foam-in-place" shipping pack. (Do not ship your unit in a wood or metal cabinet.)

4. If you provide your own shipping pack, the minimum recommended requirements for materials are as follows:

- a. 275 P.S.I. burst test, Double-Wall carton that allows for 2-inch solid Styrofoam on all six sides of unit or 3 inches of plastic bubble wrap on all six sides of unit.
- b. Securely seal the package with an adequate carton sealing tape.
- c. Do not use light boxes or "peanuts". Damage caused by poor packaging will not be covered under warranty.

Using your 'foam-in-place' shipping pack

Note: The foam-in-place packing is molded so that there is only one correct position for your product.

1. Open carton and lift center cushion leaving both end-cushions in place.

18 Warranty



UNITED STATES & CANADA

SUMMARY OF WARRANTY

Crown International, 1718 West Mishawaka Road, Elkhart, Indiana 46517-4095 U.S.A. warrants to you, the ORIGINAL PURCHASER and ANY SUBSEQUENT OWNER of each NEW Crown product, for a period of five (5) years from the date of purchase by the original purchaser (the "warranty period") that the new Crown product is free of defects in materials and workmanship. We further warrant the new Crown product regardless of the reason for failure, except as excluded in this Warranty.

**Warranty is only valid within the United States of America. For information on Warranty outside of the U.S.A, please contact your local distributor.*

ITEMS EXCLUDED FROM THIS CROWN WARRANTY

This Crown Warranty is in effect only for failure of a new Crown product which occurred within the Warranty Period. It does not cover any product which has been damaged because of any

intentional misuse, accident, negligence, or loss which is covered under any of your insurance contracts. This Crown Warranty also does not extend to the new Crown product if the serial number has been defaced, altered, or removed.

HOW TO OBTAIN WARRANTY SERVICE

You must notify us of your need for warranty service within the warranty period. All components must be shipped in a factory pack, which, if needed, may be obtained from us free of charge. Corrective action will be taken within a reasonable time of the date of receipt of the defective product by us or our authorized service center.

WHAT THE WARRANTOR WILL DO

We will remedy any defect, regardless of the reason for failure (except as excluded), by repair, replacement, or refund. We may not elect refund unless you agree, or unless we are unable to provide replacement, and repair is not practical or cannot be timely made. If a refund is elected, then you must make the defective or malfunctioning product available to us free and clear of all liens or

other encumbrances. The refund will be equal to the actual purchase price, not including interest, insurance, closing costs, and other finance charges less a reasonable depreciation on the product from the date of original purchase.

Warranty work can only be performed at our authorized service centers or at the factory. Warranty work for some products can only be performed at our factory. We will remedy the defect and ship the product from the service center or our factory within a reasonable time after receipt of the defective product at our authorized service center or our factory. All expenses in

remediating the defect, including surface shipping costs in the United States, will be borne by us. (You must bear the expense of shipping the product between any foreign country and the port of entry in the United States including the return shipment, and all taxes, duties, and other customs fees for such foreign shipments.)

DESIGN CHANGES

We reserve the right to change the design of any product from time to time without notice and with no obligation to make corresponding changes in products previously manufactured.

LEGAL REMEDIES OF PURCHASER

THIS CROWN WARRANTY GIVES YOU SPECIFIC LEGAL RIGHTS, YOU MAY ALSO HAVE OTHER RIGHTS WHICH VARY FROM STATE TO STATE. No action to enforce this Crown Warranty shall be commenced after expiration of the warranty period.

THIS STATEMENT OF WARRANTY SUPERSEDES ANY OTHERS CONTAINED IN THIS MANUAL FOR CROWN PRODUCTS. 5/11

DISCLAIMER OF CONSEQUENTIAL AND INCIDENTAL DAMAGES

YOU ARE NOT ENTITLED TO RECOVER FROM US

ANY INCIDENTAL DAMAGES RESULTING FROM

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Service Return Authorization Request

Shipping Address: HARMAN Factory Service, 1718 W. Mishawaka Rd., Elkhart, IN 46517
You may also request a service return authorization at www.crownaudio.com/support/rma

PLEASE PRINT CLEARLY

SRA #: _____ (If sending product to Crown factory service)
Model: _____ Serial Number: _____ Purchase Date: _____

PRODUCT RETURN INFORMATION

Individual or Business Name: _____

Phone #: _____ Fax #: _____ E-Mail: _____

Street Address (please, no P.O. Boxes): _____

City: _____ State/Prov: _____ Postal Code: _____ Country: _____

Nature of problem: _____

Other equipment in your system: _____

If warranty is expired, please provide method of payment. Proof of purchase may be required to validate warranty.

PAYMENT OPTIONS

I have open account payment terms. Purchase order required. PO#: _____ COD

Credit Card (Information below is required; however if you do not want to provide this information at this time, we will contact you when your unit is repaired for the information.)

Credit card information:

Type of credit card: MasterCard Visa American Express Discover

Type of credit card account: Personal/Consumer Business/Corporate

Card # _____ Exp. date: _____ *Card ID #: _____

*Card ID # is located on the back of the card following the credit card #, in the signature area. On American Express, it may be located on the front of the card. This number is required to process the charge to your account. If you do not want to provide it at this time, we will call you to obtain this number when the repair of your unit is complete.

Name on credit card: _____

Billing address of credit card: _____



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by HARMAN