

PDM

PROGRAM DELAY MANAGER

Profanity Delay Reinvented



Operator's Manual



THE TELOS ALLIANCE®

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p/n 2001-00350-000 Program Delay Manager - AES

p/n 2001-00351-000 Program Delay Manager - LIVEWIRE

TelosAlliance.com

PDM Operator's Manual

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About 25-Seven Systems

25-Seven Systems specializes in audio technologies and products that address the unique problems of radio broadcasters, networks and content providers. The company was launched in 2003 by a group of veteran broadcasters with extensive audio experience and a significant portfolio of intellectual property. 25-Seven joined The Telos Alliance in 2012. Its products are designed and built in the United States.

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Warnings

Important Safety Information

ELECTRICAL WARNING

If the equipment is used in a manner not specified by the manufacturer, the protection provided by the equipment may be impaired. The installation and servicing instructions in this manual are for use by qualified personnel only. To avoid Electric Shock, do not perform any servicing other than that contained in the operating instructions unless you are qualified to do so. Refer all servicing to trained service personnel. When using this equipment, the following safety precautions should always be followed to reduce the risk of fire, electric shock, and injury to people.

Read and understand all instructions.

Follow all warnings and instructions marked on the equipment.

Installation must be performed in accordance with all national wiring rules.

Caution: DOUBLE POLE/NEUTRAL FUSING

The PDM power supply incorporates an internal fuse. Hazardous voltages may still be present on some of the primary parts even when the fuse has blown. If fuse replacement is required, replace fuse only with same type and value for continued protection against fire.

While the power supply includes an external power switch, this product's power cord is the primary disconnect device. To prevent risk of electric shock, disconnect power cord before servicing. The outlet to which this equipment is connected must be installed near the equipment, and must always be readily accessible. The unit should not be positioned such that access to the power cord is impaired. If the unit is incorporated into a rack, an easily accessible safety disconnect device should be included in the rack design.

This equipment has an autoranging line voltage input. Ensure that the supply voltage is within the specified range.

This equipment is designed to be operated from a power source that includes a third "grounding" connection in addition to the power leads. Do not defeat this safety feature. In addition to creating a potentially hazardous situation, defeating this safety ground will prevent the internal line noise filter from functioning.

MOISTURE WARNING:

This equipment is intended for INDOOR USE ONLY and can be hazardous if immersed in water. To reduce the risk of electrical shock, do not expose this product to rain or moisture. Keep moisture away from the ventilation openings in the sides and rear of the unit. To avoid the possibility of electrical shock, do not use this equipment when you are wet. If you accidentally drop the equipment into water, do not retrieve it until you have first unplugged all cords. Do not reconnect this equipment until it has dried thoroughly. Do not shower or bathe with PDM.

VENTILATION WARNING

This equipment requires the free flow of air for adequate cooling. Do not block the ventilation openings in the sides or rear of the unit. Failure to allow proper ventilation could damage the unit or create a fire hazard. Do not place the units on a carpet, bedding, or other materials that could interfere with any panel ventilation openings.

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Updates

Features and operations of Voltair are determined largely by software. 25-Seven Systems strives to provide the most stable and feature-rich software available. We encourage you to check for software updates from time to time by visiting our website or by contacting us directly.

Feedback

We welcome feedback on any aspect of our products or this manual. In the past, many good ideas from users have made their way into software revisions or new products. Please contact us with your comments or suggestions.

We support you...

By Phone/Fax

You may reach our 24/7 Support Team in emergencies by calling +1 216-622-0247. For billing questions or other non-emergency technical questions, call +1 216-241-7225 between 9:00 AM to 5:00 PM USA Eastern Time, Monday through Friday.

By Email

Non-emergency technical support is available at Support@TelosAlliance.com.

By Web

The 25-Seven Web site has a variety of information that may be useful for product selection and support. The URL is TelosAlliance.com/25-Seven

SERVICE

You must contact Telos Alliance before returning any equipment for factory service. We will need your unit's serial number, located on the back of the unit. We will issue a return authorization number, which must be written on the exterior of your shipping container. Please do not include cables or accessories unless specifically requested by the Technical Support Engineer. Be sure to adequately insure your shipment for its replacement value. Packages without proper authorization may be refused. US customers, please contact 25-Seven Technical Support at +1-216-622-0247. All other customers should contact local representative to make arrangements for service.

Warranty



For the latest Telos Alliance warranty, visit: telosalliance.com/warranty

Register your product

Register your product today to get the full benefits of our warranty, support, and product updates. telosalliance.com/product-registration/

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Close-ups

Some parts of this manual explain how Program Delay Manager works “under the hood”. You don’t need to read these sections to operate or install the system, but they’re interesting and understanding them will help you use the system better.

Program Delay Manager at a Glance

Broadcasters have been using Profanity Delay for decades. The first schemes used analog tape loops, and delayed programming by several seconds. Operators who heard something inappropriate switched the audio feed to real-time, stopping those last few seconds from getting to the transmitter (and presumably, keeping listeners and the FCC happy).

Before the next program segment, the operator would play a jingle or promo to the transmitter, while the host talked into the tape delay. This would build up a fresh delay for the next obscenity.

Tape delays were soon replaced by trouble-free digital ones, and many of today's units can slow down the input signal to build a delay without interrupting programming. Other features may include automatic playout of filler material, a 'cough button' to let hosts speak privately to a producer while delayed audio is on the air, and a smooth ramp back to real-time when desired.

25-Seven's Program Delay Manager combines the functions of the best delay units with some unique and potentially vital features:

Automatic audio logging of every "Dump" event:

- What gets fed to the transmitter is automatically recorded on a time-stamped audio file. There's no question of what actually went on-air.
- What got dumped is *also* automatically recorded, on a time-stamped "off-air" audio file. There can't be any question of what actually took place in the studio or over your phone lines.
- On-air log files provide proof of standards for regulatory authorities. You can instantly demonstrate what people actually heard, not what they *thought* they heard.
- Off-air audio files provide proof of fairness for listeners. If you had to dump a guest comment, you can demonstrate exactly what was considered objectionable.
- Off-air audio files can also be used as extended content for less restrictive environments. Create Web promos, or let Internet listeners hear highlights of 'what couldn't go on the air!'
- Log files provide a valuable record for program management and personnel development or evaluation.
- All log files include short audio segments from before and after the event, so you can identify their context.

Automatic PD-Alerts™ and archives for management:

- Whenever an operator presses the Dump button, our patented¹ PD-Alert system can automatically send e-mail text alerts—or actual audio files—to your Program Director, Legal Department, or anyone else you designate. Or send text to the PD's and GM's cell phones *and* audio files to the Legal Department. No separate computer or operator intervention is necessary².
- Any valid e-mail address can be used. Send PD-Alerts to offsite servers, or anywhere else you want.
- The PD-Alert system is controlled by password-protected access from a computer on your network. Operators cannot change or bypass these Alerts from PDM's front panel.
- Audio files are also stored in non-volatile memory inside Program Delay Manager. They can be retrieved, with password protection, from a connected computer.
- PDM stores hundreds of megabytes worth of audio files internally. It automatically purges the oldest ones, so it constantly maintains a secure record of the most recent dump events.

And of course, the flexibility and smart design you expect from 25-Seven:

- Glitch-free time manipulation using technologies developed for 25-Seven's groundbreaking Audio Time Manager®.
- Stereo processing with CD quality, even during time manipulation. It's clean enough that you can use obscenity protection on live music programs and other events.
- Easy operation from the front panel, wired remote, producer's computer, or Axia console.
- Total flexibility: PDM can build delays without using filler material, by seamlessly stretching live audio. Or it can fill time with built-in audio playback of pre-recorded jingles or other filler material, or by triggering your automation system. Or it can use a combination of these methods. Or talent can pre-roll live material before a segment starts, with a visual count-down showing when the delay will be filled.
- Multiple Dump modes: Eliminate objectionable material seamlessly, jumping to newer material in the buffer and then automatically rebuilding the delay. Or use PDM's Overkill™ Dump mode, which plays back your custom "filler" material without affecting the delay.
- PDM can exit from delayed segments instantly, or by subtly speeding up the end of a segment until it catches up with real-time, or by playing out the delayed audio at normal tempo.

¹ US Patent 8,352,629 "Media Stream Capture, Modification, and Forwarding", January 8, 2013.

² You will need to give PDM a network connection and access to an e-mail system, since PD-Alert uses Internet e-mail. But it doesn't need a separate computer to run: all necessary functions are built into PDM.

- You can change setup and parameters any time, as your station's needs change.
- Flexible front-panel LCD dashboard with controls for time readout, audio level display, and quick parameter changes without a computer.
- Automatic synchronization of remote time displays, so talent can give accurate time checks even while audio is delayed.
- Built-in delay for Program Associated Data [PAD] and other information over IP and RS-232. Data streams are automatically synchronized to match the audio delay.
- Built-in flexibility for serial data and commands. Communicate via RS-232, RS-485, or dedicated IP Ethernet ports.
- Complete, easy to use, password-protected configuration and control from any browser. No need to navigate through menus. Use PDM's graphic user interface from any computer on your local or wide area data network.
- Flexible audio configuration:
 - XLR version includes separate stereo analog and AES/EBU inputs and outputs.
 - Livewire version for flexible audio-over-IP installation, in an Axia environment.
- Flexible remote control:
 - Use GPIO (buttons, relay closures, or TTL), secure Web interface, RS-232, IP packets, or Axia/Livewire³.
 - GPIO functions are fully assignable by the user.
 - Flexible, powerful serial language for custom, bi-directional control and status reports. Language can be supported by many automation systems and custom applications.
- Network cues can be automatically delayed to match program audio.
- Automatic synchronization to local or Network Time Server.

For more details on what PDM does and its special features, read the rest of this manual.

³ Requires Livewire version of PDM.

Basic Operation

PDM's flexibility doesn't get in the way of easy operation⁴. These functions are also explained graphically, starting on [page 16](#).



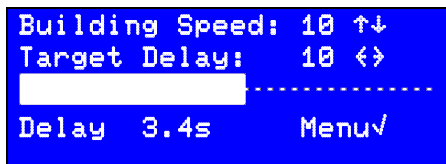
To begin a program segment...

...press the green BUILD button. It will flash while the delay is building up.

Talent should start talking immediately. Depending on how your system is set up⁵, PDM will either:

- Build a delay by expanding: PDM will send the incoming audio to the transmitter. But it'll be subtly slowed down, taking more time until the delay memory is filled.
- Build a delay by playing a station jingle, ID, or other fill material from PDM's internal memory.
- Build a delay by Pre-rolling: PDM mutes its output while you play material from another source, or while PDM signals your automation or cart replacement system to play audio.

While the delay is building, the LCD will look like this:



The ↑ ↓ ↔ and ↙ symbols mean you can use front panel buttons to change parameters or access PDM's menus.

There's more about using the front panel buttons on [page 27](#).

The horizontal bar will fill, and Delay on the bottom line of the display will count how many seconds of audio are in PDM's delay memory.

As soon as there's enough audio in memory to dump programming smoothly, the DUMP button will light.

The delay memory can hold more audio than is needed for a single Dump event. When it reaches a limit you've previously set⁶, the BUILD button's light stops flashing. The DUMP button remains lit.

4 These examples assume a normal installation: the console program output is routed through PDM before it goes to the transmitter or network feed; talent and producer are listening to the console's program output, not to the PDM's delayed output. Other installation possibilities are covered later in this manual.

5 You can change which mode gets activated easily, at any time, from the front panel LCD or remotely from a Web browser. See [page 29](#).

6 You can also change this limit at any time.

You can choose how long PDM takes to build a delay. Choose a setting based on how quickly your talent talks, how subtle you want the time manipulations to be (slower speaking styles can work with higher speed settings), and how long you're willing to wait for the delay memory to fill.

As a general guide:

Speed Setting	Minutes:seconds to build 8-second delay
7*	2:00
10	1:30
15	1:00

*factory default



To dump an objectionable comment...

...tap the red DUMP button. PDM will eliminate the last few seconds of audio that you heard in the control room, before it can get to the transmitter or network feed. How many seconds get dumped depends on how PDM is set up, and on how much delay memory it's had a chance to store.

- The size of each Dump event is determined when the unit is set up, but can be changed at any time—even while you're on the air.
- When there's enough audio in memory to smoothly Dump the full amount you've preset, the DUMP button will light⁷. This starts a few seconds after you begin a segment, or as you rebuild after an earlier Dump event.
- You can also press DUMP when it's not lit: PDM won't have enough audio stored to Dump *all* the seconds you've designated for a dump event, but it will dump whatever it has available.
- As long as the DUMP button is held down, incoming audio will be discarded.

After a Dump event, PDM begins to rebuild its memory with incoming audio, using subtle time expansion so programming is never disrupted.

⁷ For extra safety, you can also connect one of PDM's GPIOs (page 34) to keep your phone system off-air until enough delay has been built for a smooth Dump, or to light a "delay unsafe" warning for producers.

Overkill™ Mode

PDM also lets you dump incoming material by playing an internal audio file instead of collapsing and rebuilding the delay. For example, a comedy show might want to cover objectionable comments with a string of cartoon sound effects. The normal delay doesn't get depleted, and doesn't have to be rebuilt. We call this *Overkill Mode*.

- The Controls Menu ([page 29](#)) selects whether PDM uses Normal or Overkill mode when you press DUMP. You can change this setting as needed, for different program formats.
- That menu also lets you select which audio file you've loaded into PDM to play during Overkill.
- PDM's Dump Size and Delay Amount automatically adjust to the length of the selected Overkill file: for example, if you want an 8-second delay, create an 8-second file of replacement audio. After the file finishes playing, PDM's output switches back to delayed audio.



To dump a longer comment...

...tap the DUMP button again, while PDM is dumping. Each time you do, PDM will add the preset Dump Size's seconds to the current Dump event. (For example, if you've set a Dump Size of 4 seconds, tapping twice will dump 8 seconds.)

- In Normal Dump Mode, PDM obviously can't dump more seconds than it has stored in its delay memory.
- In Overkill Mode, each press of the DUMP button starts playing the Overkill file from its beginning. You can create a "stutter" effect or longer dumps by pressing the button multiple times.



PANIC MODE

If a guest continues to generate objectionable material, even after the delay is empty: press *and hold* the DUMP button. This mutes PDM's output.

Panic mode works only when PDM is set to Normal dump mode. It isn't needed for Overkill mode, since the buffer doesn't get depleted.

Cough

If talent wants to talk privately with the producer or guest...

...press and hold the blue COUGH button. While you're holding it down, PDM will continue to send delayed audio to the transmitter. But it won't store any new audio coming in at the input. Anything talent says will be heard only in the studio and control room, and won't go to air. When you let go of the COUGH button, PDM starts to rebuild the delay.

The Cough function is usually used for short comments (and of course, coughs or sneezes) that won't last longer than the Dump Size. But it continues to protect your signal during longer comments: If you hold down the COUGH button for more seconds than are stored in memory, PDM will *mute* the output. It will also flash the COUGH button to warn you of dead air.

Multi-tasking

Whenever you press the DUMP button, program audio gets dumped... as you'd expect. But PDM also does other things when you press that button:

- a) PDM starts writing a pair of audio files, one of what was broadcast, and one "off-air" of what actually took place in the studio. You have indisputable records of both what the listeners heard, and what they didn't hear because of the dump.
- b) It then adds a few seconds of extra audio to these files, taken from before and after the dump event. This makes it easier to identify the context.
- c) PDM stamps the files with the current date and time, and stores them in non-volatile memory.
- d) It sends either the complete files, or text notes that they've been created, to e-mail addresses in your PD-Alert system. Setup instructions are on page 47.



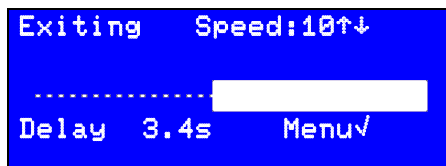
To smoothly end a delayed segment...

...press the yellow EXIT button. PDM will stop storing new audio, and will send any existing audio in its memory to the transmitter. Then it will send incoming audio directly to the transmitter in real-time.

Depending on how your system is set up:

- PDM will play any remaining audio using time compression, subtly speeding up what's in memory to rejoin real-time faster; or
- PDM will play what's in memory with no speed change⁸. Some operators call this function *roll out* or *wait and exit*.

Whichever method you choose, PDM tells you where you are in the exit process. The LCD will look like this:



When `Delay` equals 0 seconds, PDM exits Delay mode, feeding the input signal to its output.

⁸ Even if you've set the system to exit by speeding up, you can access this no-speed-change mode at any time: After you press EXIT, press and hold down COUGH, and then wait. Watch the LCD screen. When it gets close to `Delay 1.0s`, release COUGH and start talking again: PDM will smoothly transition to real-time mode.

Operating Modes

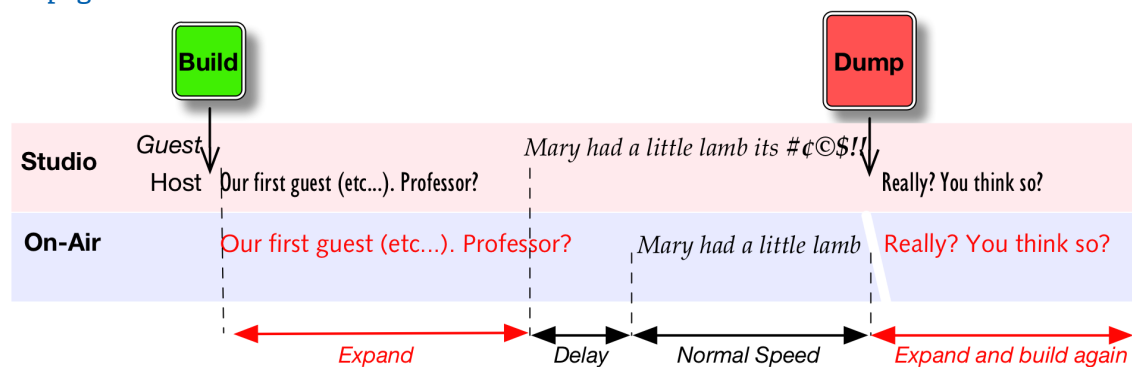
A visual note...

PDM is an advanced time-manipulation tool for radio broadcasters. But since we can't play audio in this manual, we'll use graphics to explain how PDM operates.

These drawings might also help you visualize what PDM is doing when you press its buttons. The top line (pink background) shows what's happening in the studio: this is what's heard by the talent and producer. The bottom line (blue background) shows the resulting audio that PDM sends to the transmitter or network feed. The *white slash* through the blue line, beneath the DUMP button, shows where PDM has smoothly deleted material from the on-air program.

Build by Expanding, then Dump

Build is the process of recording incoming audio to PDM's delay memory, so there's enough built up to cover the gap during a Dump event. There are other ways to build delay, shown on [page 20](#).



- Talent starts the program at the usual time, presses the BUILD button, and starts talking. PDM will subtly slow down⁹ the signal it sends to the transmitter (shown in red), while storing the real-time signal in its delay memory.

When the delay memory reaches a pre-set limit, PDM's output returns to normal pacing (shown in black). But the output is delayed by that pre-set amount, to give you protection from objectionable material.

- When talent or a producer hears something unacceptable, they press the DUMP button. PDM mutes the delayed signal before the obscenity, and instantly jumps back in memory by a pre-set number of seconds. Talent can continue talking, and the audience doesn't hear dead air.

PDM then automatically starts building up its delay again.

⁹ The degree of slowdown, delay memory limit, and Dump seconds can all be set from the front panel. So can other important parameters. Instructions start on [page 26](#).

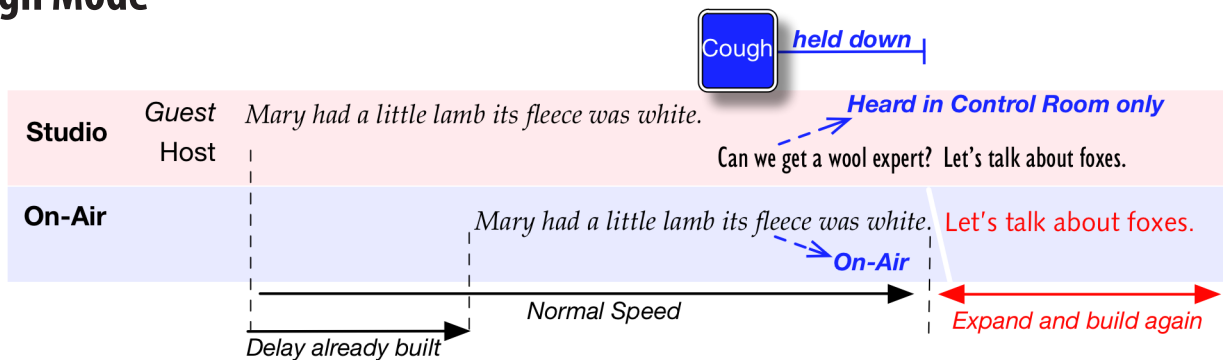
Speed changes are virtually undetectable

We use the same time manipulation algorithms we developed for 25-Seven's groundbreaking Audio Time Manager.

PDM manipulates speed by intelligently splicing individual soundwaves, so pitch isn't affected and pauses aren't unnaturally clipped. Critics have called our sound quality "superb".

For details on how this works, check www.25-Seven.com.

Cough Mode



We're into a segment, and a delay has already been built. Talent wants to give a message to the producer.

- Talent holds the COUGH button down. While it's down, the delay memory stops recording and talent is heard in the control room only. The audio already in memory (from *before* COUGH was pressed) still goes to the transmitter.

This, of course, empties the delay memory by the number of seconds COUGH is held down. When COUGH is released, PDM starts rebuilding its delay.

Another visual note...

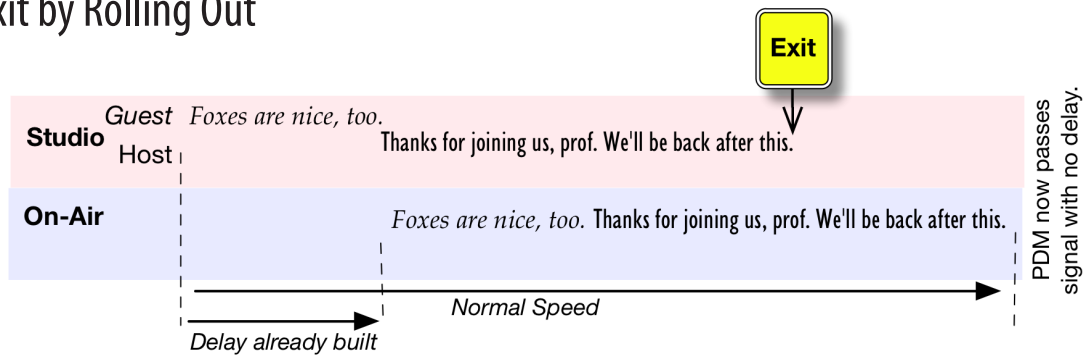
In this manual, we squeeze or stretch the type font *a lot* to illustrate how PDM adjusts audio speed. That's because we wanted to make clear what's happening.

In actual use, PDM's functions are a lot more subtle. Most listeners won't be able to tell they're in use. They're also user-adjustable, so you can fine-tune the sound to your programming and station's style.

Exit Modes

PDM can handle the end of a delayed segment, before returning to real-time programming, by *Rolling Out* or *Compressing*. Instructions for setting the mode are on [page 29](#).

Exit by Rolling Out



This is the traditional method, similar to exiting an old-fashioned tape delay:

- Talent finishes a sentence, presses EXIT, and stops speaking.
- PDM continues to play previously-recorded material at normal speed, until the delay is empty. Then PDM starts passing the incoming signal with no delay.
- Note that COUGH is disabled while exiting by rolling out.

In this mode, nothing is recorded after Talent presses exit. It's like Talent's mic gets turned off during roll-out, and then turned back on when the delay reaches zero. But it all happens automatically inside PDM.

Exit by Compression

You can use this mode to exit delayed mode without talent ever having to pause.

- Talent presses EXIT and keeps on talking. PDM keeps recording audio into memory.
- PDM subtly speeds up its output. Everything in memory plays out slightly *faster* than normal. After a short while, the output catches up with PDM's input. Then PDM drops into non-delayed mode.

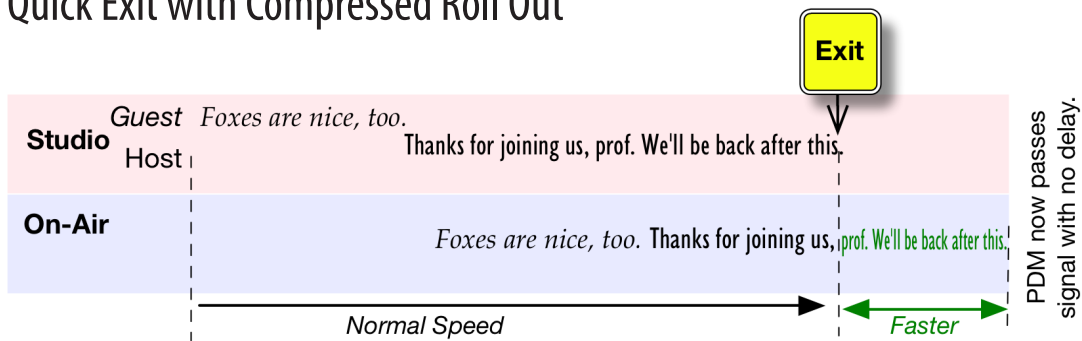
The time it takes to exit a delay using time compression depends on the size of the Delay you're using, and the Speed setting you've given PDM. As a general guide:

As a general guide:

Speed Setting	Minutes:seconds to build 8-second delay
7	1:45
10	1:10
15	:45

Choose a Speed appropriate for your talent's style of delivery, so listeners won't notice the time compression. Hosts who speak very quickly usually sound better with lower PDM Speed settings.

Quick Exit with Compressed Roll Out



This is faster than using either Compress or Roll Out modes by themselves. It assumes you've already set PDM to Compress mode.

- Talent presses EXIT and stops talking.
- PDM plays out what's in memory, and subtly speeds up its output. Whatever's left in delay memory plays out slightly faster than normal.
- Producer or network listens to PDM's output. After talent's final word, press BYPASS and go to the next program element.

Note that PDM is still recording during Compressed Roll Out, so there's a possibility that anything talent says after the segment is "over" might get passed to the transmitter. Pressing BYPASS instantly switches PDM's output to real-time and erases PDM's memory. That's why you want to press it right after talent's last word reaches the output.



Exit by Crashing Out

This instantly switches PDM to real-time mode, and empties its delay memory.

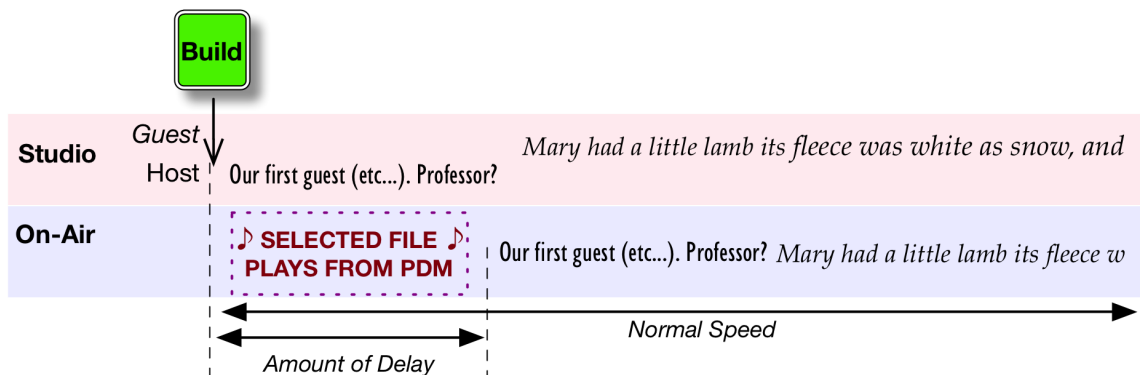
- Just press the BYPASS button.

Alternate Build Modes

You can choose how PDM will build up its delay, depending on programming needs and how you like working. This choice can be made or changed at any time by using the menu on [page 29](#).

The two other modes are *Build with Insert*, and *Build with PreRoll*.

Build by Inserting an audio file



Set Build Mode to Insert. Then when you press BUILD, PDM will play a preset audio file from its internal audio storage. Meanwhile, talent starts talking and the studio signal is sent to PDM's delay memory.

When the audio file finishes playing, PDM plays delayed audio. You can now use DUMP as needed.

You can upload the audio files PDM uses from any computer on your network. See [page 52](#).

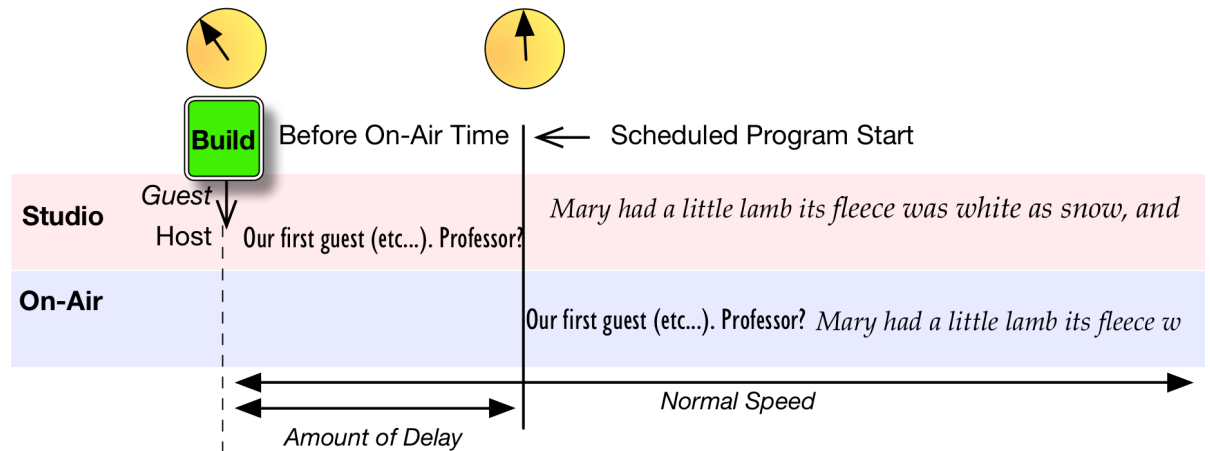
PDM can hold multiple Build files. You can have different ones for different programs, and select which one will get played from PDM's front panel or a computer. See [page 27](#).

If the selected Build file is exactly as long as the selected Delay Size, PDM will smoothly join delayed audio at normal speed when the file finishes playing.

- If the file isn't the same length as the Delay Size, PDM will still join delayed audio when the file finishes. But it will subtly speed up or slow down the output as needed, to achieve the preset Delay.

Build with PreRoll

You may want to build your delay without changing speed or playing an insert file. PreRoll Mode lets talent start their introduction *before* the scheduled program time. This is often done by studios feeding a network, rather than a local transmitter.



- Set BuildMode to PreRoll. Then, you can press BUILD and have talent start talking *earlier* than the program is supposed to air. This early start, heard in the studio only, should be the same number of seconds before airtime as you've chosen for Delay Size.

For example, if you feed network at Noon and use a 10 second delay, you would back-time and start the PreRoll ten seconds before noon. In other words, talent starts talking at 11:59:50; PDM then outputs the material starting at exactly 12:00:00.

- The LCD will start to count up with numbers and a bar graph, like this:

```

Building Speed:10  ↑↓
Target Delay:  10.0  ↔
[Progress Bar] -----
Delay  3.4s      Menu√
  
```

While PDM is counting, studio audio is sent to PDM's delay memory, and its output is muted.

When the full delay time is reached (and the solid bar fills the screen), PDM starts playing delayed audio to the air chain. If you started the PreRoll at the right time, this should be exactly the right time to start sending the program to the network.

Delay times, time manipulation, and speed selections

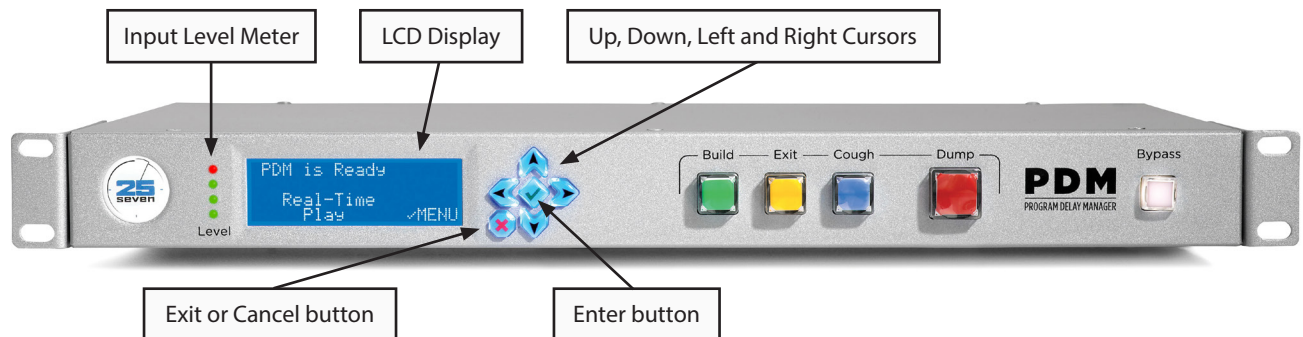
PDM uses advanced time manipulation algorithms that don't affect pitch and don't add harmonic distortion or frequency limitations. They also don't rely on deleting pauses, so they can be used on audio with continuous energy, such as musical performances or background crowds at events. PDM's algorithms excel at speech, and won't destroy your talents' delivery.

Lower speed numbers cause the least amount of tempo change, and are usually imperceptible. But they take longer to fill the delay memory, or to return to real-time.


Higher speed numbers fill the memory more quickly, but might cause a noticeable speed change on fast-paced material. The highest numbers might also cause occasional doubling of fast consonants or drumbeats while the system is building a delay.



Front Panel

Details of the rear panel appear in the Installation section, [page 63](#).



The Input Level Meter monitors the signal going in to PDM's processing circuit. At least one of its LEDs should be flashing whenever there's audio at any reasonable level. More details on this meter appear below.

The Cursor buttons let you navigate around the LCD display. In this manual, they're shown as , , , and  for Up, Down, Left and Right.

The Enter button (shown as ) is used to open menus and confirm choices. The Cancel button () exits the current menu without making any changes.

The BUILD, EXIT, COUGH, and DUMP buttons are discussed starting on [page 11](#).

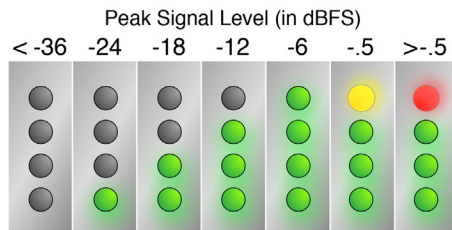
The BYPASS button is used for emergencies or when you want to take PDM out of the signal chain¹⁰. It connects PDM's analog and digital inputs to its outputs, with no processing or delay. It also clears PDM's delay memory.

PDM automatically goes into electrical bypass, connecting its XLR inputs to its matching outputs, when power is removed. Analog inputs are passed to their respective analog output; AES/EBU input is passed directly to the AES/EBU output. This is totally passive and no level adjustment or analog/digital conversion takes place.

¹⁰ Occasionally, you may need to do software updates or other system-level maintenance ([page 68](#)). Put the unit in Bypass mode, or remove it from the signal chain, before beginning these operations.

Input Level Meter

PDM's four color-coded LEDs indicate audio input level at the processor. They're connected after the analog-to-digital converter on XLR-equipped PDMs, so they can accurately monitor any input regardless of the `Input Sensitivity` setting.



- If all four LEDs are dark, the input signal is below -36 dBFS. This usually represents normal studio background noise, with no talking or music.
- As the peak input signal increases between -36 dBFS and Full Scale, successive LEDs light up.

All four LEDs will light yellow if PDM is expecting a Livewire input but not seeing a valid signal. This is usually a configuration problem. All four LEDs will also light yellow in the unlikely event of internal error, even on XLR (analog and AES/EBU) systems.

PDM's input meter shows instantaneous peaks as a diagnostic aid. It doesn't display RMS voltage or time-integrated level, and isn't a replacement for the sophisticated loudness meters found in a modern broadcast facility.

Detailed Operating Instructions

When PDM is first powered up, transport buttons will light, and the unit will be in Bypass mode. After about 10 seconds, the LCD displays this initialization screen for a few moments:

```

Welcome to
25 Seven Systems
PDM
Please wait...
  
```

Once the hardware is successfully initialized, the screen will update; the level meters will show activity if there is a valid audio input. PDM stays in Bypass mode¹¹, and a screen like this appears:

```

PDM is in Bypass
10:32:00a
Menu✓
  
```

This says the current time is 10:32 AM. PDM can display time in 12 or 24 hour modes; see page 23.

Press \diamond to enter the configuration menus.

After you turn off Bypass mode, the “ready” screen appears. This screen also appears after you exit a delay.

```

PDM is Ready
10:32:00a
Delay 0.0      Menu✓
  
```

This says PDM is processing audio, and delaying it by whatever amount is displayed on the bottom line of this screen.

- When you first enter Ready mode, PDM has no delay. This makes the audio transition seamless.
- If you press build, PDM starts to add delay and this number counts up.

If you press the BUILD button while the main screen is showing, PDM will start building a delay.

¹¹ You may also see a brief `Waiting for audio stream` message while PDM initializes its audio hardware.

General menu procedures

A few elements are common to many of PDM's screens:

Menu \downarrow means press \diamond to see a deeper menu. See **PDM's Menus**, below.

OK \downarrow means press \diamond to apply a parameter change, and return to the next higher menu. But it won't be saved permanently within PDM. See **Confirming Changes**, below.

CNCL \times means press \otimes to return to the next higher menu without making any additional changes. If you're on the Master menu, pressing \otimes exits the menu system and takes you to either the Main operating screen or the Confirmation screen (next section).

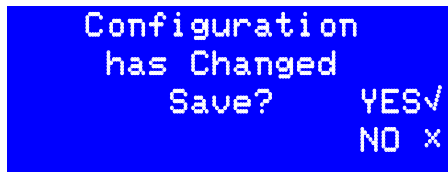
\updownarrow Chg value means use \uparrow or \downarrow to change the value highlighted on this screen

\leftrightarrow Chg field appears when a screen has more than one value that can be changed. Use \leftarrow or \rightarrow to select which value is changed by the \uparrow or \downarrow buttons. The active field will be flashing on PDM's screen.

More \uparrow and More \downarrow mean use \uparrow or \downarrow to see more menu options in this category

Confirming Changes

If you've made any parameter changes while in the menu system, you'll see a Confirmation screen like this before you return to the Master menu¹²:



Press \diamond to permanently save these changes

Press \otimes to erase these changes and restore PDM's former settings

If you don't press either button, the Confirmation screen times out after three minutes. Any changes you just made in the menu system are cancelled, and PDM returns to its normal operating screen.

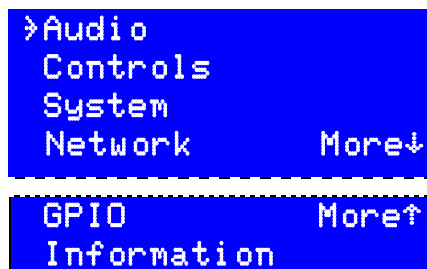
PDM's Front Panel Menus


These menus can be activated from PDM's front panel, or via its secure Web interface. For this part of the manual we'll show the LCD versions.



If you're controlling PDM from a Web browser, most of the GUI versions of the menus will be virtually identical. The Web interface also includes extra configuration options that aren't available from the front panel. These start on [page 39](#).


Master menu


If you press  while `MENU` is displayed, the screen will change to look like this¹². Here you can choose to adjust:



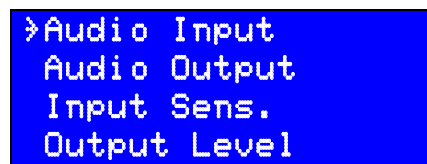
Press  to enter the Audio menu

...or use  and  to point to other menus on this screen

“More” means keep pressing  to scroll to additional options (shown in this manual under the double dotted line).

Press  to see previous options (shown above).

Audio menu



Select analog or digital inputs.

Select AES/EBU or s/pdif digital output. Analog output is always on.

Adjust the analog input level to match your studio's "0 VU"

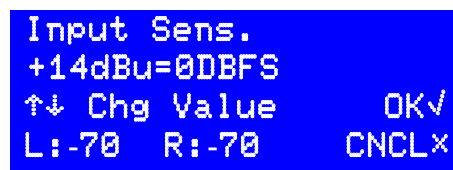
Adjust the output gain to match studio's "0 VU"

Audio menu options



Audio input Choose Analog, digital AES/EBU, or digital s/pdif (all input formats are stereo). Only one of these formats is available at a time.

Digital out Choose AES/EBU or s/pdif. This affects both the data format and the electrical specification of the digital audio output connector ([page 64](#)). The digital output is always active, even if you've chosen an analog input.

Input Sens. (Sensitivity) The LCD will look like this:



Current value is +14 dBu: this should be the absolute loudest analog signal PDM will see in your installation.

Press  or  to change the input sensitivity.

L and R show the resulting digital levels for analog signals coming into PDM. You can use these readings with an external test tone, to verify Sensitivity settings.

¹² Depending on which menu you're in, you may need to press  a few times to see the Confirmation screen.

Select a value between +20dBu and -10dBu to result in fullscale within PDM and on the digital output.

The bottom line of the display shows current Left and Right internal levels in dBFS. This is the result of the incoming analog signal, adjusted by the second line of the LCD and sent to the analog-to-digital converter. Typically, these should read -12 for a +4 dBu lineup tone.

For most installations, where 0 VU = +4 dBu, we recommend setting the analog Input sensitivity to +14dBu=0dBFS.

Analog Output Level Shown when analog or digital inputs are selected. The LCD will look like this:

```
Output Level
0DBFS=+14dBu
↑↓ Chg Value   OK✓
L:-70  R:-70   CNCLX
```

This shows how the analog output level reflects digital audio levels within PDM's processor. In this example, 0 dBFS inside PDM will cause a +14 dBu analog output. This particular setting is equivalent to the "0 VU = +4 dBu" standard, with 10 dB of headroom.

Precise Left and Right output levels are displayed in dBFS

Select a value between -10dBu and +20dBu to appear at the analog outputs, when PDM's internal digital signal level is 0 dBFS. This is usually set to match the Analog Input sensitivity, so our recommendation for most facilities would be 0dBFS = +14dBu. However, you can use other settings if you want PDM to also create analog audio gain or loss. Note that during Bypass mode, any such gain or loss is also bypassed: the output equals the input.

The analog outputs are always active, even if a digital input has been selected.

Don't confuse dBu—an analog measurement—with digital dBFS!

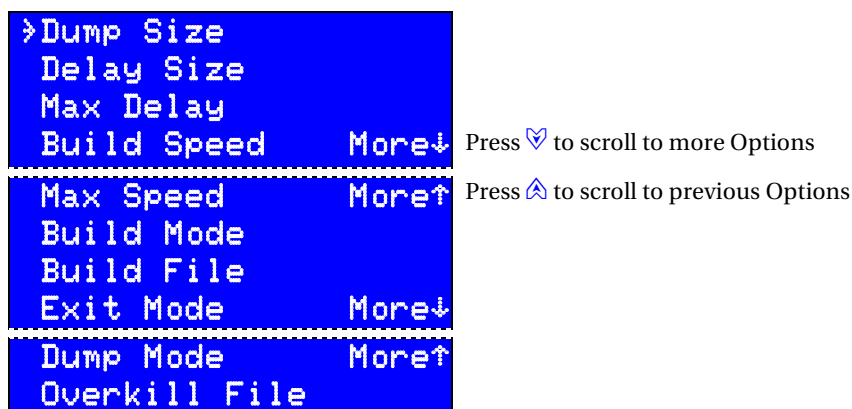
Analog dBu is a comparison to an arbitrary voltage (.775 v), and signals can be much hotter. A console's output is typically +4 dBu (1.228 v) when the meter reads 0 VU, but that's for a sinewave test signal. Peaks in voice waveforms, in real-world broadcasting, can sometimes be considerably above that... even though properly-calibrated VU meters are deliberately too slow to register them. Analog is very forgiving of these momentary peaks.

However, in the digital world, 0 dBFS (decibels referred to Full Scale) is an absolute ceiling. Some devices distort horribly or crackle when an input attempts to pass this value. PDM handles overloads differently, applying serious limiting to gracefully protect your signal (and listeners).

If your facility uses the +4 dBu standard of modern broadcast equipment, our recommended setting of +14dBu= 0dBFS provides 10 dB of headroom for protection while still maintaining an 84 dB signal-to-noise ratio. With this setup, it's unlikely that PDM will ever go into distortion or audible limiting.

If you're dealing with heavily-processed feeds you can use a lower setting; for seriously dynamic performances it may need to be higher. PDM has 94 dB dynamic range at all settings, within 2 dB of the theoretical limit for 16-bit sound.

Controls menu



Controls menu options

Dump Size Displays a screen of Dump Seconds, in .1 second increments. Use ↑ and ↓ to select how much programming PDM will delete when you press the Dump button. Choose between 1.0 and 60.0 seconds.

You can tap the Dump button multiple times for particularly long deletions. You can also tap it if you hear additional programming that needs to be dumped, even while PDM is rebuilding its delay following a previous dump. Each tap will delete the number of seconds set by Dump Size, up to the limit of how many seconds have been saved in PDM's delay memory. This delay amount is displayed on the bottom line of PDM's main screen.

Delay Size Displays a screen where you can use ↑ and ↓ to select how much delay PDM will store in memory, in .1 second increments. Choose between 1.0 seconds and the maximum set by the next menu item.

The Delay size can also be changed from *outside* the menu system, if in a Delay mode. Use ← or → to change the Delay seconds when PDM is showing a normal Delay screen (Building, Exiting, or Full).

Max Delay Use ↑ and ↓ to set a limit up to 99 seconds for the Delay Size function noted immediately above. This is to prevent operators from accidentally setting unrealistic delays.

Build Speed Sets how quickly PDM can build or dump its delay memory. Choose between 1 and 20¹³.

Max Speed Sets a limit between 5 and 20% for how quickly PDM can build or exit a delay

Build Mode Use ↑ and ↓ to select what signal appears on PDM's output (and is usually fed to the transmitter or network) while building up a delay. Choose Expand, Insert, or PreRoll. These choices are described on [page 20](#) of this manual.

13 Some adjustment options might not appear on the Master or Audio menus, depending on the hardware version you have. Axia Source and Destination menus don't apply to PDM-XLR versions. Audio connections and levels aren't used on the PDM-Livewire version.

Build File Lets you choose which jingle or other audio file saved in PDM will play out when building a delay, if **Insert** has been selected in the menu above.

Use **▲** and **▼** to select the file's name. PDM can store multiple audio files for flexibility in programming.

Use the Graphic User Interface ([page 39](#)) to upload and manage audio files in PDM's internal audio file library.

Exit Mode Use **▲** and **▼** to determine how PDM behaves when you return to non-delayed programming by pressing EXIT. Choose **Roll Out** or **Compress**; these choices are described on [page 13](#).

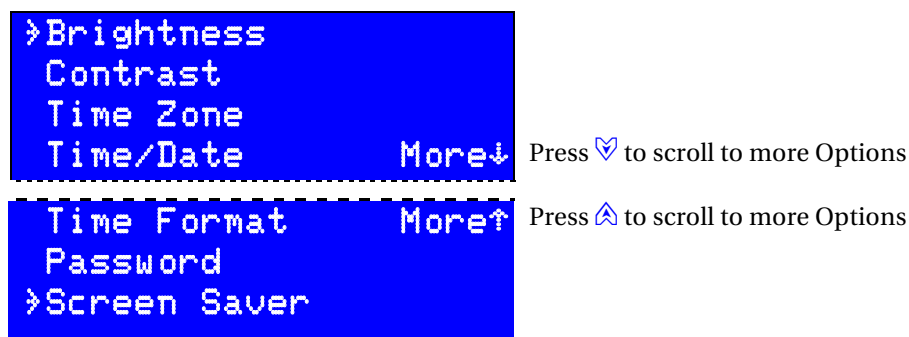
Dump Mode Lets you choose between **Dump** and **Overkill** mode.

- **Dump** is the normal mode most people are used to: it "jumps ahead in time", skipping the objectionable audio. It then rebuilds PDM's delay automatically as needed.
- **Overkill** plays a pre-selected file instead of the objectionable audio. The delay is maintained and doesn't have to be rebuilt.

Overkill File This menu lets you choose which audio file previously uploaded to PDM will play during a dump event.

- When an **Overkill File** is selected, PDM's delay and dump time automatically adjust to match the length of that file. For example, if you want a particular program to have an 8-second delay and you're using **Overkill**, create and select an **Overkill File** that runs 8 seconds.
- Note that when **Overkill** is selected as the **Dump** mode, **Delay** and **Dump** menu choices disappear from the **Control** menu since their values are automatically set.

System menu



System menu options

Brightness Select a value for the LCD display

Contrast Select a value for the LCD display

Time Zone Select a time zone when Network Time Protocol (a feature that automatically sets PDM's internal clock, described on [page 33](#)) has been selected in the **Net** menu. Choose from the four named US time zones, with or without Daylight Savings Time, or any internationally used offset from Greenwich Mean Time.

Time/Date Set the current time and date manually, when Network Time Protocol is not being used.

Why bother setting the clock? Because PDM's PD-Alert function uses time and date to identify audio log files.

Time Format Select 12 Hour (AM/PM) or 24 Hour ("Military time").

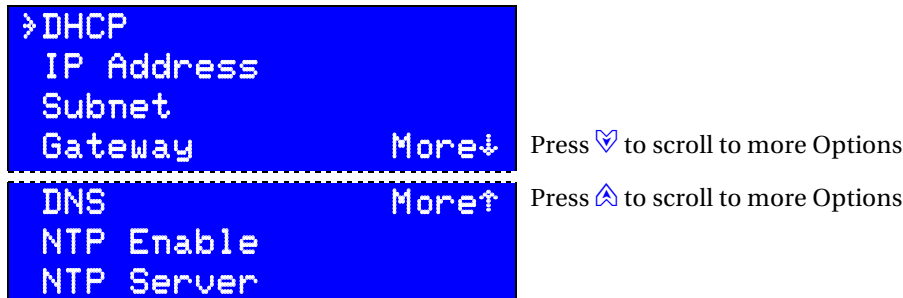
Screen Saver Use ↑ and ↓ to set how many minutes of inactivity will pass before PDM's LCD display goes dark. Or choose **Never**, if you want the screen to remain lit at all times.

Password Lets you change the password required to access the web remote control. The factory default is `pdmwebExxx`, where `xxx` is the software module number of your system. Use up/down arrows to change the character value, left/right arrows to change fields.

- Valid characters include upper and lower case letters, numbers and spaces between characters. Note that spaces at the end of a character string are removed.
- PDM passwords are designed to prevent unauthorized access via the network. They are visible from PDM's front panel menus.

Note: In previous PDM software, the System Menu also included **RS-232** and **External Clock Offset**. These functions now appear only on the Web interface ([page 44](#)).

Network menu¹⁴



Menu may be slightly different for Livewire-equipped PDMs. See [page 68](#).

Network menu options

DHCP Choose `enabled` or `disabled`.

Use `Enable` if your network router is set up for DHCP address assignment. When DHCP is enabled, your unit's IP address, Subnet, DNS Server, and Gateway are supplied by the router. `Enabled` is the factory default.

- Disabling, pressing `DONE`, then re-enabling and pressing `DONE` will ask your system's router to reset PDM's IP address. It may take a few moments before a new address is registered.
- Depending on your network's router, after you enable and confirm a change to DHCP, you may need to power cycle PDM to get an address.
- PDM responds to Pings across your network. You can use this to determine if an address is valid and the network is working properly.

Fields in the following menus "roll over" between their limits of 0 and 255. If the `0` is showing, press [down arrow] to jump to 255; if 255 is showing, press [up arrow] to jump to 0.

These first four menu options appear only if DHCP is `Disabled`. Otherwise, the information is supplied automatically by your router.

IP Address When manually setting network addresses, use [left arrow] and [right arrow] to change fields; [up arrow] or [down arrow] to set values.

Subnet mask In most cases, this should be set to `255.255.255.0`.

DNS Server DNS stands for "Domain Name Server". This number is supplied by your Internet Service Provider [ISP] or IT department. Just about every modern computer, including PDM, uses one of these servers to access the Internet.

Gateway The IP address of your LAN's Gateway system or router that connects PDM to the Internet, also supplied by your ISP or IT department. It's needed so the PD-Alert system can send its e-mail to servers outside your local area network.

¹⁴ The +14 dBu sensitivity in the figure would make these meters read -10 dBFS with a standard +4 dBu line level test signal. In other words, the setting means a +14 dBu signal would read 0 dBFS—the maximum allowable digital signal—and standard line level is 10 dB less than that, or -10 dBFS.

NTP Enables or disables PDM's ability to update its time-of-day clock automatically from a Network Time Protocol server¹⁵ (see "About NTP" below). If enabled, PDM's clock cannot be manually set.

NTP Server Enter the IP address of the network time server. This menu item appears only if NTP is enabled.

Network troubleshooting

PDM's Internet services (such as PD-Alert e-mails, Network Time Protocol, and Web remote control) require proper Internet Protocol addressing. In many installations this is provided automatically by your station's data router, when you choose DHCP (page 32). However, some data networks don't support DHCP, or require fixed IP addresses. If that is the case, you will have to manually enter the IP Address, Subnet Mask, Gateway, and DNS numbers.

About NTP

Network Time Protocol servers supply accurate time-of-day and other data over the Internet. PDM can use them to automatically set its internal clock. Your facility might also have its own NTP server.

NTP servers are often identified by a domain name—a familiar-sounding ".com" or ".gov" URL—rather than their numeric IP address. PDM requires numeric IP addresses.

You'll find a list of US servers, with both their domain names and their IP addresses, at the National Institute of Standards and Technology's website¹⁶. International time server domain names are listed at www.ntp.org. Please respect the access policies listed by the server provider and IT department. PDM users should never select a Primary server for NTP updating; use Pool or Secondary servers instead.

The best way we have found to test whether NTP services are working is to manually offset your PDM clock by a couple of minutes, then enable NTP. If your network connections are working and your NTP address is valid, the system clock should quickly acquire the correct time.

If your Information screen shows an IP address of 0.0.0.0 and DHCP has been selected, it is likely the router isn't supplying a valid address. Power cycling PDM may be necessary to activate DHCP. In some cases you may need to power cycle or reset the router as well. Note that router resets can temporarily disrupt the entire data network.

PDM's Network Time Protocol system will also report potential network issues while it is establishing a connection with the time server. These messages appear on the information screen, and are described on page 38.

15 The factory default is 7, which takes about two minutes to fully build an eight-second delay. That's subtle enough to be imperceptible to most listeners. There's a discussion of Build Speed on page 11.

16 The list itself is currently at <http://tf.nist.gov/tf-cgi/servers.cgi>. If NIST has moved it from that location, use the search box on their home page, www.nist.gov, to find "Internet Time Service".

GPIO Menu (General Purpose Input/Output)

```

>GPIO Enable
GPIO Inputs
GPIO Outputs
GPIO Test
  
```




This menu lets you program the parallel inputs and outputs for remote control.

GPIO Enable Select Enabled to activate General Purpose Input / Output control, a parallel remote control mode accessed through the female DB-25 jack on the rear panel. Select Disabled if you have a remote connected to this jack but want to temporarily turn it off.

GPIO wiring connections are through a rear-panel DB-25 jack; details are covered on [page 65](#). GPIO can also be connected directly to consoles and nodes through Livewire, if you have the proper hardware; details are on [page 59](#).

PDM's GPIO system doesn't use noisy relays and won't be tripped by normal static impulses. You can leave it Enabled even if there is no remote control plugged in.

GPIO Inputs and **GPIO Outputs**

The GPIO system has to be configured to determine which PDM functions appear at which pins on the rear-panel DB-25 jack. To configure GPIO, use  or  to select **In_s** or **Out_s** and press . The LCD will look like this:

```

GPIO Inputs:
In1=Dump
↑↓ Chg Value      OK✓
←→ Chg Field     CNCL✗
  
```

Each of the 8 Inputs or outputs (e.g., In1, In2, Out1, Out2, etc) can be assigned to a PDM function, as listed on the following chart. The same function can be assigned to multiple inputs or outputs, if you require two or more isolated circuits for it.

As a general guide:

GPIO FUNCTION	INPUT SETTING*	OUTPUT SETTING†
Nothing Assigned	None	None
Build button	Build	BuildTrig
Exit button	Exit	
Cough button	Cough	
Dump button	Dump	DumpTrig
Bypass button	Bypass	Bypass
Cue point #1 [¶]	Cue1	Cue1
Cue point #2 [¶]	Cue2	Cue2
...etc (up to 8 Cue points)	etc	etc
Flag point #1 [¶]	Flag1	Flag2
Flag point #2 [¶]	Flag2	Flag2
...etc (up to 8 Flag points)	etc	etc
Pause: temporarily suspend delay build	Pause	
Utility point #1 [¶]	Util1	Util1
Utility point #2 [¶]	Util2	Util2
...etc (up to 4 Util points)	etc	etc
Delay memory: pulse on Full		FullTrig
Delay memory: pulse on Empty		EmptyTrig
Build: mimic lamp state		BuildLamp
Exit: mimic lamp state		ExitLamp
Dump: mimic lamp state		DumpLamp
Cough: mimic lamp state		CoughLamp

Building delay in progress		Building
Exiting delay in progress		Exiting
Delay currently full		DelayFull
Delay safe: dump increment achieved		DelaySafe
Delay unsafe: dump increment not achieved		DelayUnsafe
Delay currently empty		DelayEmpty
Output muted (when Dump or Cough are held down and there is no audio in delay)		Muted

Important notes about the table

* — Where an input function has the same name as a PDM front-panel button, the input behaves just like that button.

For example, a momentary pulse on a GPIO `Build` input will start building the delay, just as if you'd tapped the BUILD button.

† — There are multiple outputs for some PDM functions:

- If an output setting ends with `Trig`, PDM generates a single output pulse when that state is reached. This can be handy for interfacing with automation systems. Output pulse length can range from 100 to 1000 ms and can be set in the Configuration page “Hardware GPIO” Section. Default value is 250 ms.
- If an output setting ends with `Lamp`, PDM's GPIO output mimics the front panel button's lamp. It stays low when the lamp is on, high when the lamp is off, and cycles when the lamp is flashing. This can be handy for building custom remotes for talent and producers.
- `Building`, `Exiting`, `DelayFull`, `DumpReady`, `DelayEmpty`, and `Muted` indicate PDM's current operational state, and are held low for as long as these states exist.
 - `DumpReady` indicates when the Dump Size (page 27) has been reached, and a single dump is now possible. It can be connected to your phone system or console to absolutely block incoming calls until they are safe.
 - `DelayFull` can be used to tell talent or producers that the maximum delay has been reached. In most setups, this will mean that multiple or extended dumps are also possible.

¶ — “Cue” and “Flag” let you automatically delay the signals from satellites, automation systems, and producers' control panels. This keeps the signals in sync with appropriate program audio at all times.

Connect this kind of contact closure or logic pulse to any PDM GPIO input that you've designated as a **Cue** or **Flag**: when a signal is received, it's linked to the incoming audio sample. It stays with this sample, no matter how much time manipulation is being applied. You can then pass the mark to other devices by assigning an appropriate GPIO output.

Cues and Flags usually indicate important programming events such as commercials or news breakaways. PDM preserves their timing as much as possible, even if the audio that originally carried them has been dumped.

If you dump audio that contained a Cue or Flag, PDM will send the appropriate signal during the "splice" between older and newer material in its output stream.

- **Cue** marks generate a 250 ms trigger (or other 100 to 1000 ms value, as configured) when they're reached.
 - For example, if the network provides a contact closure when it's time for a local break, connect it to an input that has **Cue1** assigned. No matter how much delay is applied, PDM will generate a **Cue1** output when that sample is reached... which you can use to trigger other equipment to play the break.
- **Flag** marks record a continuous logic input. Their output reflects that logic state when the appropriate audio sample reaches PDM's output.
 - For example, you can use a Flag to activate a skimmer or logging recorder. Connect a mic's tally signal to the input, and use its output to keep the recorder on whenever the mic is on the air... regardless of how much delay is in use.

φ — Utility GPIOs 1-4 allow you to instantly trigger an input closure and fire it on an output closure without any delay. This feature is useful for implementing or diagnosing end-to-end relay logic when connecting PDM to other devices.

TEST

The GPIO test screen lets you verify that GPIO connections and external remotes are properly wired.

```

12345678
Ins: 11111111
Outs: 00000000
↑↓↔ Change      CNCLX
    
```

When you leave the GPIO test screen, the DB-25 outputs are all reset to 0.

Information

Choosing **Information** from the Master menu lets you scroll through screens displaying PDM's current Audio, Network, and Time/Date settings; NTP Status; network MAC address, and internal temperature. As a general guideline, PDM should run within 20° Fahrenheit of ambient room temperature¹⁷. It also provides the unit's unique software module number, and the current version numbers of its front panel firmware and system software. You'll need this information if you call 25-Seven for support.

The NTP Status display shows one of three messages, indicating the current connection with a remote time server:

- **Search** PDM is seeking a path to the remote NTP server. If this message persists, it may indicate that the NTP server address you entered isn't valid.
- **Query** PDM has contacted the remote server and calibrating its internal clock to match. This may take a few minutes depending on the connection and how much adjustment is required.
- **Sync** PDM's clock is now synchronized with the remote server's.

¹⁷ There's more about heat control in the section on Installation, page 62.

Graphical User Interface and Web Remote Control

You can access PDM from a Web browser in any computer on the data network, if you have the proper password.

Logging in to PDM from a Connected Computer

- Launch any standard Web browser, on any Windows, Macintosh, or Linux computer on your network¹.
- Set the browser to `http://nnn.nn.n.nn`, where the n's are PDM's IP address. You can read that address from the **Information** menu on PDM's front panel.
- When it connects, you'll be asked for a User Name and Password

Use **pdmweb** as the **user name**, and **pdmwebBnnn** as the **password**. (*Bnnn* is the software unique Module Code, which you'll find on the **Information** screen. *pdmwebBnnn* is the factory default password. You can change or view your password from the front panel in the **System / Password** menu ([page 31](#)). Passwords are case sensitive.

If it doesn't connect, you may have a problem with your network connections, the version of Flash installed in your browser, or your system's configuration¹⁸.

After successfully connecting, PDM will draw its Graphical User Interface on your Browser's screen.

Port Forwarding & GUI Communication

PDM's web server connects at *http port 80*. Once you log in and the front panel or configuration pages are loaded, embedded Adobe Flash movies in our GUI let your browser control the PDM via *port 5444*. Both these ports must be open and available for the GUI to function.

If you are trying to connect to your PDM over a WAN or across the Internet, you will need to use a NAT router¹⁹ to route ports 80 and 5444 to your PDM's local IP address.

If your facility has multiple studios, you may want to control two or more PDMs on the same network using a single external IP address. You can do this with your NAT router along with PDM's built-in port offset feature. The key is to forward *pairs* of port numbers.

¹⁸ This may be a setup issue in PDM's menus or your local computer, a network wiring or configuration problem, or even something as simple as entering the IP address wrong in your browser. PDM responds to Ping commands as an aid for debugging.

¹⁹ Network Address Translation Router. Most routers have the ability to forward port requests from specific IP addresses. Some routers call this feature Virtual Servers. See your router's instruction manual for details.

For example, say you have two PDM's: *PDM-A* and *PDM-B*:

- At your router, configure the ports so inbound browser traffic on port 81 forwards to port 80 on PDM-A, while port 5445 forwards to port 5444 on PDM-A. Users seeking access to PDM-A from the outside would use the IP address plus port number, and enter 10.11.12.100:81 in their browser.
- For your second unit, configure inbound browser traffic on port 82 to forward to port 80 on PDM-B, while port 5446 forwards to port 5444 on PDM-B. Users seeking access to PDM-B from the outside would use the IP address and that port number, such as 10.11.12.100:82.

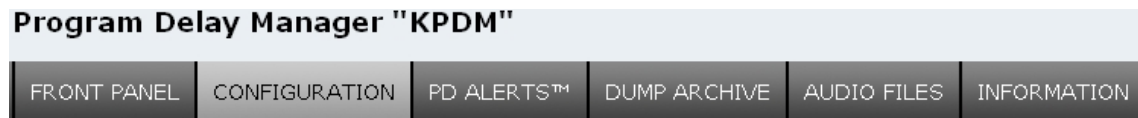
You should see a pattern here: Ports 80 and 5444 are paired. When inbound browser traffic comes in at port x , the Flash movie control port will automatically be set to $x - 80 + 5444$. You don't need to specify Flash ports, just which port your router uses to talk to PDM: if you set that and associated firewall permissions properly at the router, our software does the rest.

For more information contact customer support.

Selecting GUI Pages

PDM's GUI is organized by pages, to make it easier for you to find particular functions.

You'll see a row of tabs across the top of your browser's screen, as shown in this screenshot. Click a tab to select which page you'd like to work on.



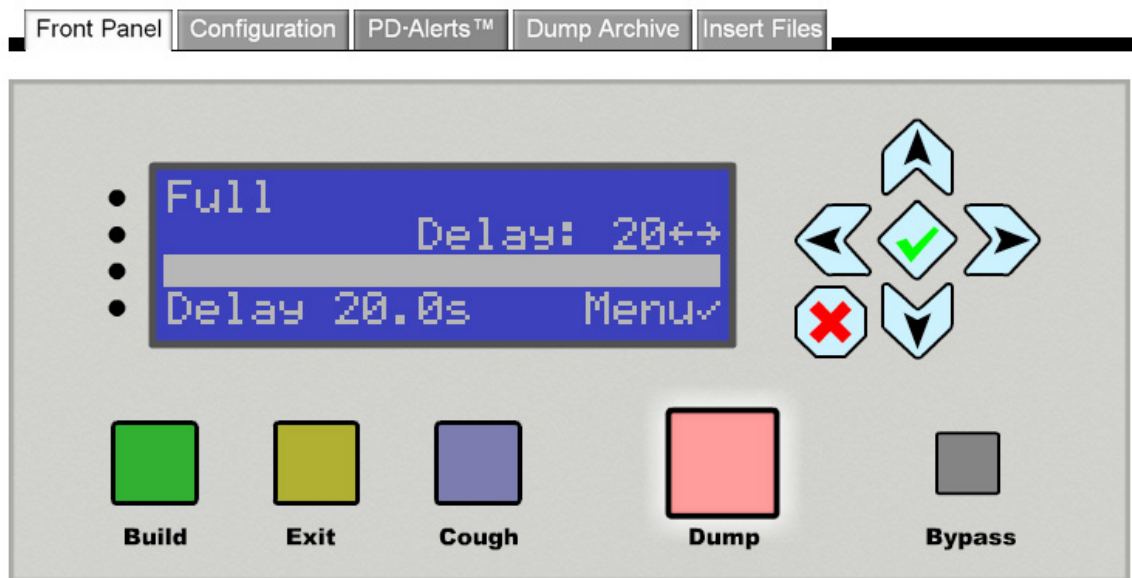
The top line displays the machine name of the connected PDM. You can change this name by clicking the tab.

Front Panel GUI Page (Web Remote Control)

You can operate PDM from this page, the same way you would from its hardware front panel, or use the pane to monitor LEDs and buttons.

When you first log in, this page appears by default. If you're on another page, you can get here by clicking the tab at the top of the GUI.

Program Delay Manager, TalkStudioA



PDM Front Panel. (It's actually much larger in your browser, approximately 720 x 330 pixels.)



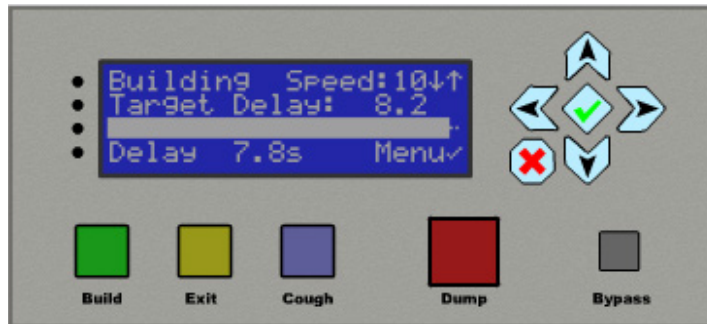
When using the GUI to control a PDM in another studio, always check to make sure the unit isn't in use! **Remember: a remotely controlled PDM, in Ready mode, may actually be feeding audio to your program bus.** Changing its settings remotely might disrupt programming.

This page is fully interactive. On-screen buttons can be clicked with your mouse; the text displays whatever is on PDM's LCD at the time; Audio level LEDs and button lamps will flash in sync with the hardware front panel.

You can use this screen to configure PDM, by mouse-clicking the cursor keys and navigating through menus. But it's a lot faster to do this from the screen (page 35).

Front Panel GUI “Mini-Window”

Click the mini front panel link directly under the BUILD and EXIT buttons, to get a Mini-Window. This opens a new window in your browser, with a small but fully functional front panel interface.



Mini-Window. (Nominal size is 370 x 165 pixels, but you can resize it on your browser's screen.)

This small version of the front panel can be resized by dragging on its lower right corner. It can also be kept on-screen while other browser windows or desktop applications are running, without taking up lots of screen space.

Multiple Mini-Windows

You can control more than one PDM simultaneously from the same computer. All you need to know is each unit's IP address, its user name, and its password ([page 39](#)).

- Log in to any connected PDM, and click the second link at the bottom of the **FRONT PANEL** screen on the GUI:
Click [here](#) to open multiple front panel windows
- Another Browser tab will appear, with a dialog where you can enter the IP address of any other PDM accessible to your network:

Add PDM Front Panel

PDM web address

Address of PDM

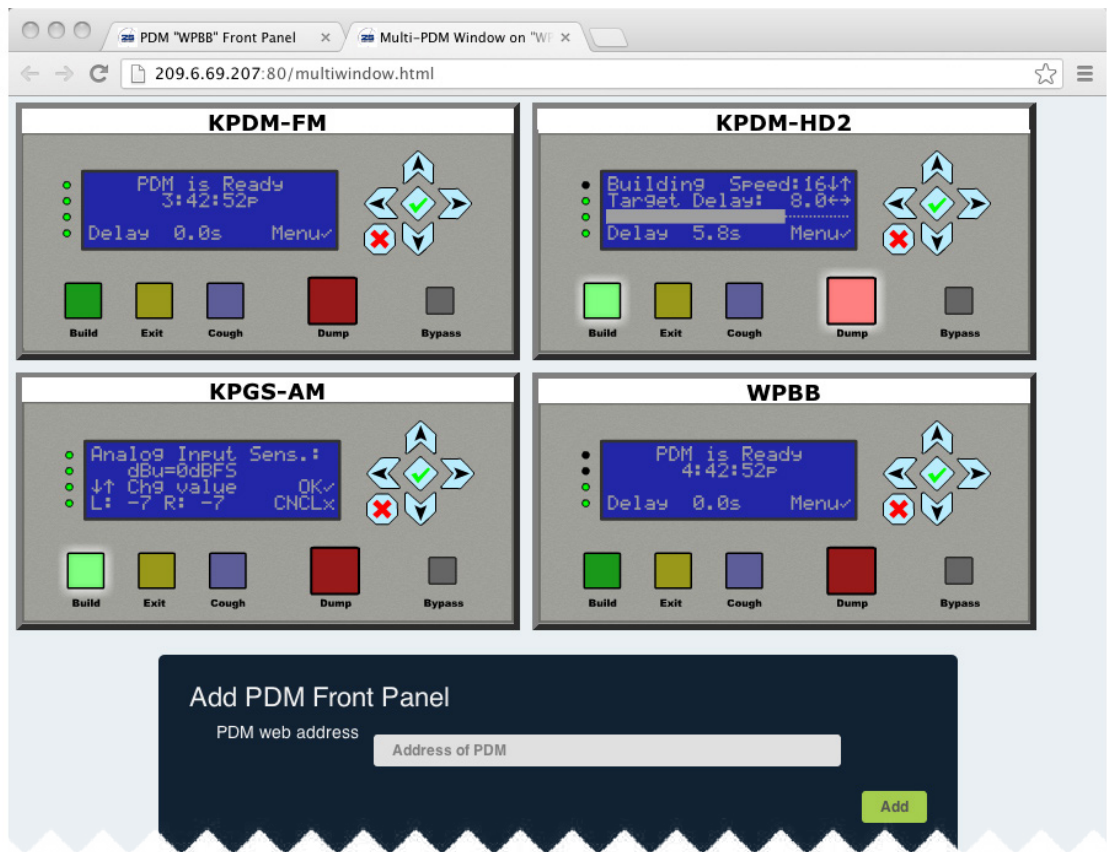
Add

- The first IP address you see will be for the PDM you're already connected to. Click **Add**, and a small version of its front panel will appear on the screen. The **Add PDM Front Panel** dialog will shift to make room for it.
- Add IP addresses of additional PDMs as desired, log in to those machines with user name and password when requested, and their front panels will also appear. You can have as many as twenty front panels on the screen, each interactively controlling its PDM.

- PDM names (as set in the Configuration page) will be displayed at the top of each front panel, so you can identify specific units.

These small front panels are displayed in the order you entered them. To change their order, or delete particular machines:

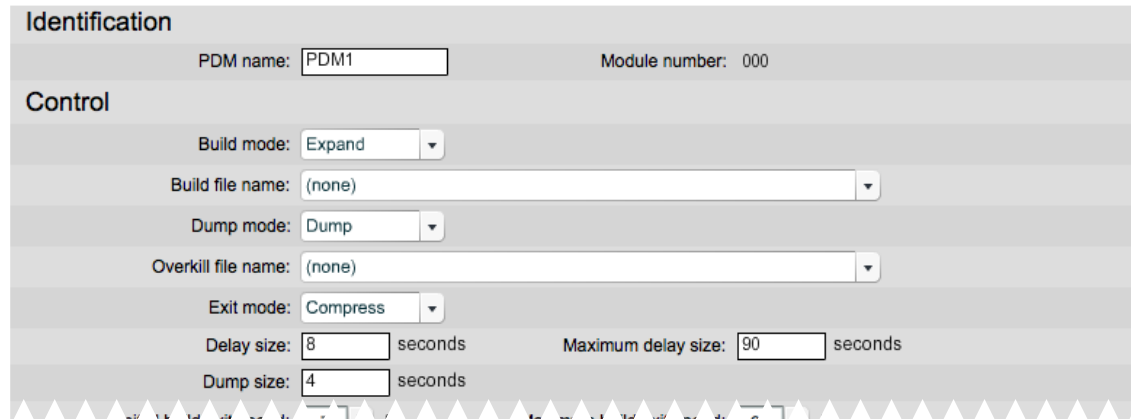
- close the multi-PDM Window in your browser,
- click **multiple front panel windows** again,
- enter desired addresses in the new order.



Up to 20 PDM front panels can be displayed and controlled simultaneously, in a single browser window.

Configuration page

Click on the **CONFIGURATION** tab at the top of the GUI, and your browser changes to look like this:



The screenshot shows the Configuration page with two main sections: Identification and Control. The Identification section includes fields for PDM name (PDM1) and Module number (000). The Control section includes dropdown menus for Build mode (Expand), Build file name (none), Dump mode (Dump), and Overkill file name (none). It also includes input fields for Delay size (8 seconds), Maximum delay size (90 seconds), and Dump size (4 seconds).

Identification	
PDM name:	<input type="text" value="PDM1"/>
Module number:	<input type="text" value="000"/>

Control	
Build mode:	<input type="text" value="Expand"/>
Build file name:	<input type="text" value="(none)"/>
Dump mode:	<input type="text" value="Dump"/>
Overkill file name:	<input type="text" value="(none)"/>
Exit mode:	<input type="text" value="Compress"/>
Delay size:	<input type="text" value="8"/> seconds
Maximum delay size:	<input type="text" value="90"/> seconds
Dump size:	<input type="text" value="4"/> seconds

Top part of Configuration page; there's more on the next page of this manual.

Bottom part of Configuration page, continued from previous picture.

PDMs with Axia / Livewire software will have a few additional entries; see [page 71](#)

Duplip size: 14 Sx 301 JS
 Initial build/exit speed: 7 % Maximum build/exit speed: 20 %

Hardware GPIO

Hardware GPIO: Enabled Pulse length: 250 milliseconds

Inputs: In 1 Build In 2 Exit In 3 Cough In 4 Dump
 In 5 Bypass In 6 Cue1 In 7 Cue2 In 8 Cue3

Outputs: Out 1 BuildTrig Out 2 FullTrig Out 3 DumpTrig Out 4 EmptyTrig
 Out 5 Cue1 Out 6 Cue2 Out 7 Cue3 Out 8 Cue4

Delayed Data Streams

Data Stream A Input: Disabled
 (remote address: remote port: 0)

Data Stream A Output: Disabled
 (remote address: remote port: 0)

Data Stream B Input: Disabled
 (remote address: remote port: 0)

Data Stream B Output: Disabled
 (remote address: remote port: 0)

Other I/O

Control via net port 5443: Enabled
 Control via RS-232: Enabled RS-232 rate: 57600 baud
 RS-485A port mode: Disabled Clock offset: 0 seconds ahead
 AES/EBU input rate: 48 kHz

PD-Alert

Mail server address:
 Mail server username:
 Mail server password:
 Sender: (email address)
 PD-Alert context: 5 seconds before and after dump
 PD-Alert (attached audio) emails: (space-separated list)
 Attached audio file format: Uncompressed WAV
 PD-Alert (text) emails: (space-separated list)
 Retain dump files: forever (or until space runs out)
 Test PD-Alerts: (sends test email to every configured address)

Front Panel

Screen saver: 2 minutes Brightness: 75 Contrast: 40

Save Changes

Changes that you enter on this page don't take effect until you press the Save button on the bottom.

- If someone enters changes at PDM's front panel while this GUI is open, the changes won't be automatically reflected on your networked computer. You have to Refresh the page to see any the changes. To avoid confusion, don't try to make configuration changes on both the GUI and PDM's front panel at the same time.

Identification shows the hardware and software of this PDM. You can set a PDM name for easy identification of this hardware. This name will also be used to create log file names and PD-Alert e-mails.

- If you have more than one PDM, make sure they have different names.
- Names can be 19 characters, including numbers or upper/lower case letters, and dashes -. Spaces and special characters (e.g.: & % @) may not be used.



Control and Hardware GPIO let you view and change settings that are on the matching menus of PDM's front panel (as described starting on [page 29](#)). Since the GUI has drop-down menu selections, you'll probably find this faster than navigating through the front panel.

Pulse length lets you adjust the output pulse duration of contact closures between 100 and 1000 ms. This is useful when connecting to some older equipment requiring longer pulse inputs. Factory Default is 250 ms.

Delayed Data Streams lets you keep Program Associated Data and text or automation cues in sync with the audio at all times. Two separate simultaneous streams are supported, and there are flexible options for sources and outputs. They're discussed fully on [page 61](#).

Other I/O lets you:

- Enable bi-directional serial commands over network port 5443.
- Configure the rear panel RS-232 port to serve as a bi-directional serial control. You may also disable the port from this menu. If you are using the RS-232 port to delay data, you must disable this selection.
- Set parameters for an external slave clock output over RS-485. If you do, you can choose to have the clock automatically compensate for PDM's current delay, as well as for delays in other equipment.

You can use the Clock Offset setting to add *extra* compensation for other delays in your air chain, like those added by HD broadcasting. Use  and  to adjust that compensation in one-tenth second increments. If you don't want to compensate for other delays, leave it set to 00.0.

- Set the baud rate for the rear panel RS-232 serial lines. Choose standard values between 1200 and 115200 baud.

PDM's serial command language and data delay capability is described fully on [page 55](#).

What time is it?

If you give talent or producers a clock that's been slaved to PDM's output, they'll be able to announce time checks and manage the program while staying in sync with the rest of the world. For example, if PDM's delay is 8 seconds, a connected remote clock will show 11:30:00 when the time is actually 11:29:52. Whatever gets said at that moment will come out of PDM exactly at 11:30.

PDM's slave output stays in sync no matter what you do. For example, if you DUMP and delete 8 seconds of programming, the studio clock instantly jumps back 8 seconds. If PDM is building a delay by slowing the output, the clock slows down to match.

PD Alert configures PDM's automatic dump-event logging and alerting function. This important feature and its settings are discussed separately, below.

Front Panel mimics the front panel menu settings on PDM's LCD for visual brightness and contrast.

Save Changes The Save button changes PDM's internal settings to match whatever you've entered on this page. Revert cancels your recent entries, replacing them with whatever had been previously set in PDM.

- Whenever you make changes on the configuration page, the title of that selection appears in **bold** to indicate a change is pending, and the Save and Revert buttons will flash.
- When you successfully Save those changes, the selection title reverts to normal text, The Save and Revert buttons stop flashing, and a confirmation dialog appears on screen. You must click OK on this dialog before you can make other changes.
- If you type an invalid value or setting, that field appears in red to indicate an error.

The PD-Alert System

When a user presses the DUMP button, PDM can automatically e-mail text alerts—or actual audio files—to anyone you designate. This is our patented PD-Alert feature. For security reasons, operators can't change the PD-Alert setup from the device's front panel: you must log into the Graphical User Interface from a network computer, using a password.

PD-Alert uses standard Internet e-mail protocols, so it has to be configured like a standard e-mail account. You'll need an outgoing server address, a valid user name, and the password for that user's account. This is the same information you'd use setting up Microsoft's Outlook Express, Mozilla's Thunderbird, or any other e-mail program on a desktop computer²⁰.

You can use any existing e-mail account. But most stations find it more efficient to establish an account specifically for PDM.

²⁰ PDM doesn't receive e-mails; it only sends them. So it doesn't need an incoming "POP" server. Also, PD-Alert does not support the SASL [Simple Authentication and Security Layer] protocol, so you won't be able to use web mail services like Gmail.

Mail server address: Enter a valid SMTP server address, such as the one where you normally send outgoing e-mail. A typical address may look like smtp.kpdm.com. The default mail server port is 25. If your mail server is on a port other than 25, specify it by typing a colon and the port number; for example, smtp.kpdm.com:587

Mail server username: Enter a valid e-mail account user name, *without the @ sign and domain name*. A typical entry may look like gsteadman or PDMStudio1.

Mail server password: Enter the password associated with that user name. This password is required by your SMTP server. If you use internal mail servers that don't require authentication, leave the mail server password and username blank.

Sender: Enter the complete email address for that user name (for example, gsteadman@kpdm.com). Even though PDM can't receive replies at that address, mail servers use this data to verify messages are legitimate.

PD-Alert context sets an amount of additional audio, from before and after each Dump event, which gets saved at the start and end of PDM's log files. Choose enough additional time to help you identify a Dump event's context. Usually five or six seconds is plenty, but if you are running very long program delays, set context to the maximum value.

PD-Alert (attached audio) e-mails: Enter the e-mail addresses of people or servers who should receive PD-Alerts *with attached audio files*. (Use the PD-Alert (text) field, below, for users who don't want to receive audio files. Those users can still access audio files from within PDM.)

- If you want to enter more than one "attached audio" recipient, separate each address with a space.

Attached audio file format: Use this drop-down menu to choose audio quality and data compression for e-mailed log files. Our compression is playable on most modern computers, smartphones, and other portable data devices.

- Uncompressed WAV: highest quality but also the largest files; may be too large to handle conveniently on mobile devices.
- High Quality MP3: Roughly equivalent to FM Radio, sent at about 160 Kbps.
- Voice Quality MP3: Roughly equivalent to AM Radio, at 40 Kbps.

Examples of File Sizes for PDM Compression Settings²¹

Quality	On-Air File	Off-Air File
Uncompressed WAV	1.7 MB	3.1 MB
High Quality MP3	120 KB	212 KB
Voice Quality MP3	60 KB	94 KB

No matter which compression format you choose for e-mail audio files, PDM always keeps its internally stored files as full-resolution, uncompressed .WAV.

²¹ Assuming PDM is set for 8 second delay, 8 second dump increment, and 5 seconds each of before- and after-context.

PD Alert files are typically generated at 44.1 kHz. If your facility prefers a 48 kHz digital sample rate, select “48 kHz” as the AES/EBU input rate under Other I/O; this keeps PD Alert files and time calculations accurate. This selection is visible only if AES digital input is selected as the source.

PD-Alert (text) e-mails: Use this field to enter e-mail addresses for people who should get text-only PD-Alerts; these are simple text notifications of when a Dump has occurred. They’re designed for users who don’t want to worry about large audio files sent to their Smartphone or PDA. These users can manually retrieve log files from PDM if they need them.

- If you want to enter more than one “text-only” recipient, separate each address with a space.

Retain dump files: PDM purges old audio files to make room for new files when space is needed in its internal storage. There’s a *lot* of storage, so dump files can hang around for a long time. If your station’s policy is to delete air-checks on a regular schedule, use this setting to choose 30, 60, or 90 days. PDM will automatically delete dump files as they reach that age.

Test PD-Alerts: After you’ve entered e-mail addresses, use this feature to verify that PDM is set up properly. A test message will be sent to the addresses in both PD-Alert e-mails fields.

- If one of the e-mail addresses is invalid²², your ISP will relay an error message. You can check for these messages by logging into the mail account you set up for PDM, from any normal e-mail program.

The Save button changes PDM’s internal settings to match whatever you’ve entered on this page. Revert cancels your recent entries, replacing them with whatever had been previously set in PDM.

If someone enters changes at PDM’s front panel while the GUI is open, the changes won’t be reflected on your computer until you use your browser’s Refresh feature. We suggest never trying to make configuration changes on both the GUI and PDM’s front panel at the same time; doing so can give unexpected results.

Note that unless the unit is in BYPASS mode, PD-Alerts are generated whenever the Dump button is pressed. This happens even if PDM’s delay is currently at zero... but in that case, the dump file will contain only the PD-Alert context audio as specified on this page, from before and after when DUMP was pressed.

²² This sometimes happens because of a typing error, or if an address is obsolete.

PD-Alerts page

Click on the **PD ALERTS™** tab, and the GUI will display the PD-Alerts e-mail log shown below.

The log lists PDM's recent attempts to send PD-Alert e-mails. If a message has been successfully sent, it displays each recipient's address, the type of PD-Alert sent, and the size of the e-mail including any attached audio files.

The Date column includes the date and time that Alerts were sent. These entries are clickable: if you click an underlined date and time, PDM will display more details on the event, including links to the OnAir and OffAir audio dump files.

PD-Alert™ email log
This is a list of the notifications sent for the most recent dumps.

Date	To	Type	Size (bytes)
11Mar2013 15:34:12	gsteadman@25-seven.com	Text	1127
11Mar2013 11:44:09	gsteadman@25-seven.com	Text	1126
08Mar2013 23:36:03	gsteadman@25-seven.com	Text	1120
04Mar2013 12:11:48	gsteadman@25-seven.com	Text	1118

If PDM can't send e-mail because of a network or setup error, it will display a message about the problem.

Log File Lock Box

Files generated by the PD-Alert system can be copied to your computer, and from there can be saved, forwarded or deleted like any other computer file.

Dump files in PDM's internal storage cannot be edited or deleted manually. That's for security. There is no command to do this either in the GUI or on PDM's front panel.

- Dumped files “stick around” as long as possible, until PDM's internal storage gets full. Then older ones are erased to make room for newer ones. PDM's internal storage holds *lots* of Dump files (about 700 MB).
- If your station's policy is to flush airchecks after a specific time, select a shorter period in the `Retain dump files` field (page 49).

Until a file has been deleted by the system for either of these reasons, it stays in the system and is available for review. Of course, you should also back up important ones elsewhere, by copying them to a separate system or permanent media. That's just good computing practice.

Dump Archive page

Clicking on the **DUMP ARCHIVE** tab displays a list of all the Dump files stored within PDM, with the most recent ones on top. Files are built in pairs, one for the studio or telephone audio that got deleted (OffAir) and one for what listeners actually heard (OnAir). Regardless of any compression settings on the PD-Alert page, these files are always stored within PDM as uncompressed full-resolution .wav.

If you click on any File name, PDM will play its audio through your browser. You can also transfer these files to your computer by right-clicking (Windows) or control-clicking (Mac) on the file name, and using your browser's "Save Linked File" command.

Dumped Audio Clip Archive
This is a list of the most recently dumped **PD-Alert™** audio clips.

File name	File size	Date
KPDM_21Aug2009_08:43:27_OffAir.wav	2.02 MB	21Aug2009 08:43:27
KPDM_21Aug2009_08:43:27_OnAir.wav	1.02 MB	21Aug2009 08:43:27
KPDM_21Aug2009_08:40:26_OffAir.wav	2.04 MB	21Aug2009 08:40:26
KPDM_21Aug2009_08:40:26_OnAir.wav	1.05 MB	21Aug2009 08:40:26

Part of a Dump Archive page

OnAir and OffAir files may not exceed 120 seconds in length. For the few installations running very long delays, log file audio will start with the pre-context audio, plus as much of the delay buffer as will fit in the remaining seconds.

What Gets Saved

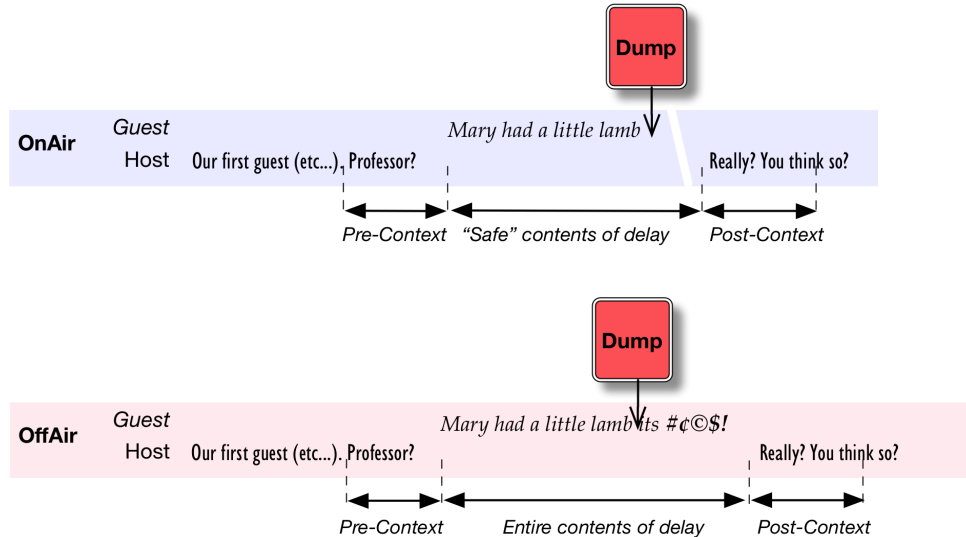
For each Dump event, PDM saves separate *OnAir* and *OffAir* audio files. Each pair is kept as full-resolution .wav for highest quality, even though you can choose to have it compressed before being e-mailed. Both files include a few seconds²³ of audio from before and after the dump event, so you can determine the material's context and easily compare files.

The *OffAir* file includes both the "safe" audio from before you pressed DUMP, and the objectionable material itself. So unless you're using Overkill Mode and covering dumped material with internal audio of the same length, *OffAir* will be longer than the *OnAir* file²⁴.

²³ Select how many Context seconds get saved on the Configuration page, described above.

²⁴ In the unlikely event that the system has zero delay when you press DUMP and you're not using Overkill, both files will be identical and consist only of the before- and after- context.

These graphics illustrate the two files²⁵:



Audio Files page

Click the **AUDIO FILES** tab to manage which audio files will be played by PDM during Insert Build Mode or Overkill™ Dump Mode. From this page, you can Delete, Rename, or Upload files. File names are links to the actual file, so you can listen to the files by clicking their names.

Build and Overkill Audio File Library
 This is a list of audio files available to build delays in Insert Build Mode, or to overwrite using Overkill Dump mode.

File name	File size	Sample rate	Date	Controls
8-sec-cartoon-sfx.wav	1.36 MB	44100 Hz	13Mar2013 13:29:31	Delete Rename
pdm_demo_insert.wav	1.39 MB	44100 Hz	13Mar2013 13:44:02	

Total file size: 2.75 megabytes
Choose a file to upload: no file selected

The file `pdm_demo_insert.wav` is included for testing, and doesn't have Delete or Rename buttons.

²⁵ OnAir files are an exact copy of what gets sent to the transmitter. So if PDM was slowing down the audio after a dump to rebuild a delay, you'll hear the effects of that expansion

- Uploaded files should be 16 bit .wav format, stereo or dual-channel mono with no data compression.
- PDM can't change the playback volume of internal files. This is so operators don't inadvertently affect your signature sound. Please make sure that files you upload are already adjusted to your station's standards.
- The file sample rate should match PDM's internal sample clock:
 - If your PDM is using only its analog inputs, it will run at 44.1 kHz and your files should be at that rate.
 - If your PDM is fed with digital audio through its AES/EBU connection, it will run at that audio's sample rate. This happens even if you're processing signals from the analog inputs, to keep the AES/EBU outputs synchronized with the rest of your studio.

If you're using Livewire, files should be Axia's native sample rate of 48 kHz.

- Maximum Insert file length is 6MB (about 35 seconds of audio at 44.1 kHz).
- PDM can store a total of 150 MB worth of Build and Overkill audio files²⁶.

The buttons are self-explanatory. If you click on an underlined file name, it will play through your browser's sound output.

²⁶ That can be literally hundreds of files, probably more than any station will ever need. To protect your own sanity when using PDM's navigation or selection screens, we recommend deleting files that are no longer in use.

Information Page

Click the **INFORMATION** tab to view information about the connected PDM. This read-only page shows configuration settings (audio, networking, and time format) along with the internal temperature and 25-Seven's identification numbers.

Note that these settings may be changed only from the PDM front panel, not via the configuration page. The page does not update automatically; refresh your browser if you need to see recent changes.

Information	
This is a read-only list of configuration and status settings. Editable items on this list may only be changed from the front panel.	
Audio	
Audio Input	Analog
Digital Output Mode	AES/EBU
Analog input sensitivity	0 dBu = 0 dBFS
Analog output level	0 dBFS = 8 dBu
Network	
MAC address	00:1f:f2:06:95:99
DHCP	disabled
IP address	192.168.1.12
Network mask	255.255.255.0
Gateway	192.168.1.1
DNS server	192.168.1.1
NTP	enabled
NTP server	216.229.0.179
NTP status	synchronized
Delayed Data Streams	
Stream A Input	Disabled
Stream A Output	Disabled
Stream B Input	Disabled
Stream B Output	Disabled
System	
Time zone	EST/EDT
Time format	12 hour
Internal temperature	90°F 32°C
Platform	JET
Firmware versions	
PDM	2.5.1
AIO	4.10.20.1
IOB	1.2.1
LSP	1.4
WEB	2.3
AXIA	1.0.8.6
UTIL	1.2

Information page. (It's a single tall column on your screen, shown as two pieces here to save space.)

Serial / Parallel Remote control

In addition to IP based remote control, PDM supports traditional control systems for maximum flexibility. All can be used simultaneously.

Parallel (GPIO or “Contact Closure”) Control

There are eight inputs and eight outputs for remote control on a rear panel female DB-25 connector. Each can be assigned to PDM functions through a configuration screen, described on [page 33](#). Multiple inputs or outputs can be assigned to the same function if you need isolated circuits. Inputs and outputs are opto-isolated, active low; wiring details are on [page 65](#).

Serial Control

PDM’s serial control options let you build custom interfaces, such as for an on-air or logging systems. You can access PDM’s internal control language via RS-232 serial, a network Web browser, or both at the same time.

The comprehensive language uses abbreviations to query delay depth and internal temperature, access all GPIO functions, and continuously report PDM’s operating status.

Standard serial or IP-based connections

Data and control lines appear on a male DB-25 connector in both RS-232 and RS-485 formats, as well as over a network.

Enable RS-232 and set its protocols in the Web interface’s **CONFIGURATION** screen ([page 44](#)). You can wire a custom cable ([page 67](#)), or use a standard DB-25 to DB-9 adapter and a null modem. You can use the RS-232 port either for control or to delay data. You can’t set it up for both functions simultaneously.

Commands and queries can also be transmitted as IP packets over network port 5443, if you enable this function on the Configuration screen.

- Once port 5443 is enabled, it remains accessible to the entire network until specifically closed. There is no further password checking. Use a firewall, or keep your PDM on a dedicated (closed) network, for protection.

Serial language structure

PDM's serial remote control language is designed to be easily implemented from any terminal program, macros in most automation systems, or computer programming language. Requests and responses are standard ASCII characters, based on plain English, using familiar PDM "front panel" designations whenever possible.

Requests, or messages from the controller to PDM, consist of a number of words, followed by a Line Feed character²⁷ <LF>. Messages are case-sensitive.

Responses are replies from PDM back to the controller. They consist of a single `Result` character, any appropriate message, and then <LF>. The Result character may be:

- ! Indicates command has been received and acted upon
- ? Indicates command isn't understood, or can't be acted upon

Output Events can be sent by PDM whenever there's a change in status, such as memory being filled or buttons being pressed. They're provided so you can design custom serial interfaces to other equipment. Use the `enable` command (page 59) to specify which events generate a message.

Request syntax

Requests consist of a *command* (something for PDM to do), usually followed by *arguments* (what it should act on), and then the <LF>.

Commands include:

- down** Start the event described by <argument>
- up** End the event described by <argument>
- trigger** Momentary activation of <argument>, same as tapping a button²⁸.
- get** Returns status of <variable>
- help** Returns hints about PDM's serial language
- enable** Turns on Output Event reporting for a specific PDM condition
- disable** Turns off reporting for a specific condition

Commands are case-sensitive and must be all lower-case.

Arguments include:

Build, Exit, Cough, Dump, Bypass, Cue1 through Cue8, Flag1 through Flag8

Arguments act the same as the identically-named GPIO inputs, described starting on page 33.

You can `enable` or `disable` Output Event reporting for just about every PDM function. A complete list of valid arguments appears on page 59.

²⁷ ASCII 10 (decimal) or 0X0A (hex). Note that Carriage Return (ASCII 13) is ignored by the system.

²⁸ `trigger <argument> <LF>` works the same as sending "`down <argument> <LF> up <same argument> <LF>`". But it sure is easier.

Arguments are also case-sensitive. They use internal capital letters, and must be sent exactly as shown in this manual.

Variables include:

Depth: current audio delay length in seconds and tenths

PeakInput: peak input level, both channels, in dBFS

PeakOutput: peak output level, both channels, in dBFS

TemperatureC or **TemperatureF:** current internal temperature in whole degrees Celsius or Fahrenheit, depending on the request.

Output Event messages sent by PDM are @NameOfEvent[=1 or =0], followed by <LF>.

NameOfEvent can be any event reported by the GPIO. If an event is momentary (such as a button press) the Boolean does not appear.

=1 or **=0** appear only if the event is an ongoing state, such as whether the system is currently in Bypass Mode. These numbers do not appear for momentary events, such as button presses or Cues.

A typical Output Event might look like:

@DelaySafe=1<LF> Means *Delay Safe* is now on²⁹

²⁹ That is, PDM's memory now has enough audio to cover at least one complete dump.

Examples

Note that requests are case sensitive.

Remote Request	PDM Response	Notes
down Build<LF>	!<LF>	PDM starts building the delay
up Cough<LF>	!<LF>	PDM releases the COUGH button
trigger Dump<LF>	!<LF>	PDM starts a DUMP event
help<LF>	!down up (etc) <LF>	* See note below
help down<LF>	! (text) <LF>	* See note below
get Depth<LF>	!Depth=7.5	Current delay is 7.5 seconds
get PeakInput<LF>	!PeakInput= -19	Current audio input peak level is -19 dBFS
	@DelayFull=1	**Delay has been fully built
	@EmptyTrig=1	**Exit completed, Empty trigger asserted
	@DelaySafe=1	**Dump increment achieved, delay is now safe
	@Bypass=0	**PDM removed from Bypass mode
DUMP GUEST!!!<LF>	?unknown command DUMP<LF>	PDM doesn't understand this instruction, so no action is taken

* — Help does not require an <argument>.

- If you send **help<LF>** by itself, it returns a list of valid commands.
- If you specify any command as the argument, it returns a brief explanation along with a list the command's arguments.

For example, if you send **help down<LF>**, PDM replies

**!down: send Press signal for one or more space-separated events
(None Build Exit Cough Dump Bypass Cuel ...etc to... Flag8)**

** — Output Event Messages (preceded by @) only appear after they've been specifically enabled for a type of event. See next section.

Output Event Messages

You can use PDM's *Output Event messages* to build elaborate custom remote controls. Any event that can trigger a GPIO output can also be sent serially.

Each type of output event must be specifically enabled before PDM will report it. This way, you can tell PDM to report only those events that are important to your installation.

Messages are sent via RS-232 and over Ethernet. When PDM first boots, or when a socket connection is made to port 5443, a *System Active* message is sent:

@Welcome to PDM xxxx³⁰<LF>. You can use this as a handshake, or as a signal to start your own routine for enabling those Output Events you want reported.

Events are enabled or disabled on a per-connection basis. You must send `enable` messages each time you establish a new connection. Otherwise, PDM defaults to having all of its event messages disabled.

enable

This command turns on PDM's ability to report specific events. Use the form `enable EventName<LF>`, where `EventName` is a change in PDM's current status. For example, `enable DumpTrig<LF>` tells PDM to send a message whenever DUMP has been activated.

- You can put two or more `EventNames` in the same `enable` command. Separate them by spaces. For example, `enable DumpTrig Muted<LF>` tells PDM to report any time either DUMP or MUTE has been activated.
- Use `All` for `EventName` as a shortcut, when you want to enable messaging for every PDM event.
- When PDM receives any valid `enable` command, it responds `!<LF>`.
- If a function is active when you enable its message, PDM reports its state immediately. It then sends an updated report when the state changes.
- For example, if you send `enable Muted<LF>` and PDM is already Muted, PDM immediately responds `@Muted=1`. Then, when Muting is turned off, PDM will send `@Muted=0`.

`EventNames` used in `enable` commands can include:

Building PDM is building up a delay
BuildTrig BUILD button has been pressed
Bypass PDM is in bypass state
Cue1...Cue8 Cue point *n* has been reached
DelayEmpty Memory is empty
DelayFull Delay is now full

³⁰ xxxx is the serial number of that specific PDM. You can use this to verify the proper system is being accessed in facilities with multiple units.

DelaySafe Dump increment has been reached in memory

DelayUnsafe Memory contains some audio, but not full Dump increment

DumpTrig DUMP button has been pressed

EmptyTrig Memory has just been completely emptied

Exiting PDM is exiting a delay

Flag1...Flag8 Flag point *n* has been reached

FullTrig Memory has just been completely filled

Muted PDM output is muted

BuildLamp BUILD button is flashing*

ExitLamp EXIT button is flashing*

DumpLamp DUMP button is flashing*

CoughLamp COUGH button is flashing*

All Enables event reporting for all of the above

***Lamp** functions mimic the lamp state, and report =1 or =0 every time a front panel button flashes on or off.

disable

This command turns off reporting of one or more output events. It follows similar syntax.

- **disable DumpTrig<LF>** tells PDM to stop reporting that DUMP has been activated.
- **disable DumpTrig Muted<LF>** turns off reporting for both functions.
- **disable ALL<LF>** turns off any message reporting.
- PDM responds to valid disable commands with **!<LF>**.

disable affects only the serial messages. Parallel GPIO functions on the DB-25 aren't changed, and stay as you set them on the front panel or via Web interface.

Data Delays

PDM can delay two data streams such as PAD (Program Associated Data), to keep it in sync with audio. This way, you can make sure “now playing” or other information on RDS receivers or web streams lines up with program audio. When this feature is turned on, incoming data is stored for exactly the same amount of time that PDM is delaying audio, and then sent out.

As audio delay is slowly built or collapsed to return to real-time, each data delay is automatically changed to match. If PDM is put into Bypass mode, data passes straight through.

If you press DUMP while Data Delay is on, stored data gets sent out immediately. This guarantees that important information isn’t lost when audio has been removed.

There are two separate data delays, Data Stream A and Data Stream B, so you can handle both internal cueing information and PAD for listeners. Inputs and outputs for each can be RS-232 or IP; IP inputs and outputs can have different addresses and ports, and can function as client or server. You can mix these types for maximum flexibility.

These choices all appear on the CONFIGURATION page, in the Delayed Data Streams section (page 46). Only one input can feed data at a time, but multiple devices can be listening to delayed data on the outputs.

Two drop down menus lets you choose the stream inputs.

- Select Disabled, RS-232 port, Local network port 5445 (for Stream A) or 5446 (for Stream B) or Remote Machine.
- If you’ve selected Remote machine, supply the IP address and port number of that machine in the address field below.

Outputs have the same choices, except they use Local network port 5447 for Stream A and 5448 for Stream B.

If you choose RS-232 port for delayed data, remember that it’s bi-directional. Data will be sent on the same RS-232 connector as it’s received: if you have one device generating data and another one listening, you’ll need a Y cable to split the TX and RX signals. This also applies if you’ve set RS-232 for data.

RS-232 can support only one function at a time. If you choose it for a data stream input, don’t enable it for the other data stream’s input... unless you want to carry the same data to different outputs. And if you’re using it for a data stream, don’t enable it for remote control.

Installation

PDM fits a standard 19" equipment rack, 1 unit high by 12" deep (allow additional depth for the connectors).

While PDM doesn't generate much heat, we recommend at least one unit of blank or perforated rack panel separating it from any heat-generating devices above or below it. Do not block the fan or intake vents on the rear of the unit.

- You can check PDM's internal temperature using front panel menus ([page 38](#)), or via web interface ([page 44](#)). As a general guideline, PDM should be within 20° Fahrenheit (about 7° C) of ambient room temperature
- PDM's cooling fan is its only moving part, and must move freely to keep PDM in normal operating range. If temperature readings appear high, check the fan and air intake vent immediately.
- We recommend you check the cooling fan annually, to make sure it's moving freely and that dust hasn't built up.

The internal power supply will self-adapt to any voltage between 100 – 240 v 50/60 Hz. It consumes 1.8 A maximum; typical draw is about 32 watt @ 120VAC.

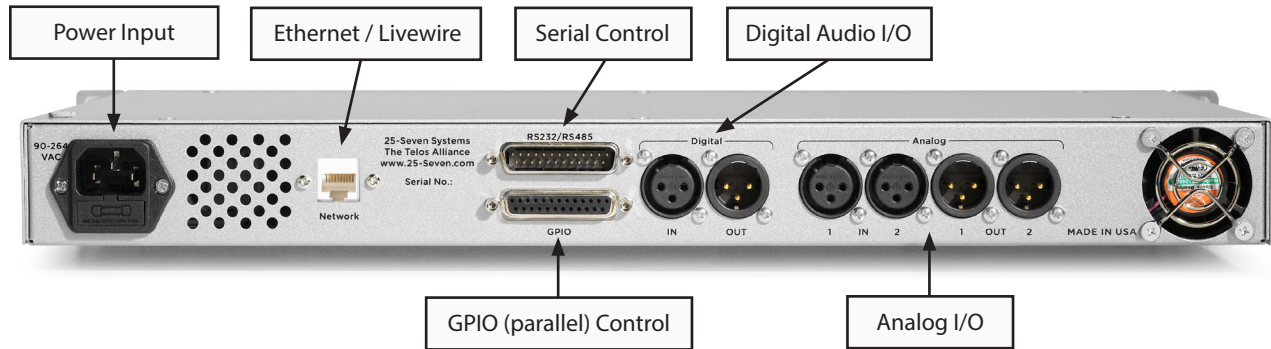
We recommend that PDM be connected to a UPS (uninterruptible power supply), to prevent loss of audio in case of a power disruption. This is good advice for any mission-critical electrical device.

XLR versions of PDM are normally fed by the Program output of a talk studio. Their output gets sent to the air chain, master control, or network as appropriate. The talent and producer should listen to the console's Program output, rather than the delayed PDM output. Other setups may be appropriate if you're using routing switchers or an automation system.

Axia Livewire versions of PDM appear as a normal Source and Destination within Axia's setup screens. See the Livewire setup section in this manual for more information.

If your station uses Arbitron PPM, place your PDM unit *before* their encoder in the signal path. This will ensure PPM encoding is not affected when audio is dumped, or when delays are being built or exited.

Rear Panel



Power Input

Standard IEC configuration self-adapts to any voltage between 100 – 240 volts AC, 50 or 60 Hz.

Ethernet / Livewire

Connect to a standard 10/100Base-T network connection. All PDM models use this port for remote control via web browser, PD-Alerts, serial remote control over IP, and synchronization to a network time server. If connected to the Internet, it should be behind a hardware firewall or on a VPN.

In additional, Axia-Livewire equipped models use this port for Livewire audio input and output. Livewire operation requires special software; Contact 25-Seven or your dealer for upgrade details.

Digital Input

When set to AES/EBU via the **Audio** menu, this input conforms to IEC 958 Professional (5v p-p, 110Ω balanced) on XLR connectors.

When set to s/pdif, the voltage and impedance switches to IEC 958 Consumer (.5v p-p, 75Ω unbalanced): connect signal to pin 2 and shield to pins 1 and 3.

The digital input jack on XLR-equipped PDMs is active at all times, even if you've selected **analog** for the audio input. This provides the most flexible operation. But it can have unexpected effects if you connect the wrong kind of signal:

- Whenever there's a valid digital input signal at 44.1 kHz or 48 kHz, PDM locks its internal sample clock to it. This assures high quality and full compatibility with other equipment in the facility, so you can use PDM as an analog-to-digital bridge. But it also means PDM internal processing can be affected if the digital input isn't stable, *and this affects both PDM's analog and digital outputs!*
- **Do not connect, remove, or change the signal on PDM's digital input jack while it's on the air!** This applies even if PDM is processing an *analog* input at the time, and if you're using only PDM's analog outputs.

- When PDM first powers up, it looks for a signal on its digital input jack. If it sees one at 48 kHz, it uses that sample rate to compute delay and ramping times. If it sees one at 44.1 kHz—or doesn't see a valid digital signal at all—it uses 44.1 kHz.
- **If you're processing a 48 kHz input signal, be sure PDM's digital input has a 48 kHz signal before turning the unit on. Or power-cycle PDM once the signal is available.** This assures that PDM's delay and timing displays remain accurate.

Under extreme circumstances, an intermittent digital input can cause erratic audio behavior. If this happens, power-cycle PDM.

Digital Output

The digital output is always active on XLR-equipped units, letting PDM act as your digital-to-analog converter: feed it an analog input signal, and use the digital output.

When set to AES/EBU via the **Audio** menu, this output conforms to IEC 958 Professional (5V p-p, 110Ω balanced) on XLR connectors. Output is 16-bit PCM linear.

When set to s/pdif, the voltage and impedance switches to IEC 958 Consumer (.5V p-p, 75Ω unbalanced): connect signal to pin 2 and shield to pin 3 only. Do not connect pin 1 for s/pdif operation.

If there is no signal on the digital input jack, PDM provides its own low-jitter clock at a 44.1 kHz sample rate. When there is a valid 44.1 or 48 kHz signal at the digital input jack—even if analog inputs are selected—the internal clock and digital output are locked to it. But see the warning about digital clocking above.

Analog Inputs

Information about input and output levels is on [page 27](#). Unbalanced operation is not recommended.

These are electronically balanced XLR female, pin 2 hot, with a load of 20kΩ: this makes them compatible with all modern electronically-balanced outputs. If fed from a transformer-balanced output, we recommend bridging a 680Ω resistor between pins 2 and 3.

Analog Outputs

These are electronically balanced XLR male, pin 2 hot, designed to feed a load of 600Ω or greater.

PDM uses advanced algorithms to process signals in true stereo, with no smearing or variation in the stereo field during time compression. So while the analog outputs (and analog inputs) are labeled CH1 and CH2 on the back panel, we assume you'll be using them for the left and right channels of a stereo pair and that's how we've written this documentation.

However, you can also use PDM for dual, identical processing of two mono signals. There's no cross-talk.

Bypass Relays

Analog and digital XLR inputs and outputs are wired to bypass relays. In the event of a power failure, incoming audio is connected directly to the outputs. PDM powers up in bypass.

Always use Bypass mode when installing software updates. You can trigger this mode with the front panel BYPASS button or the web interface.

GPIO

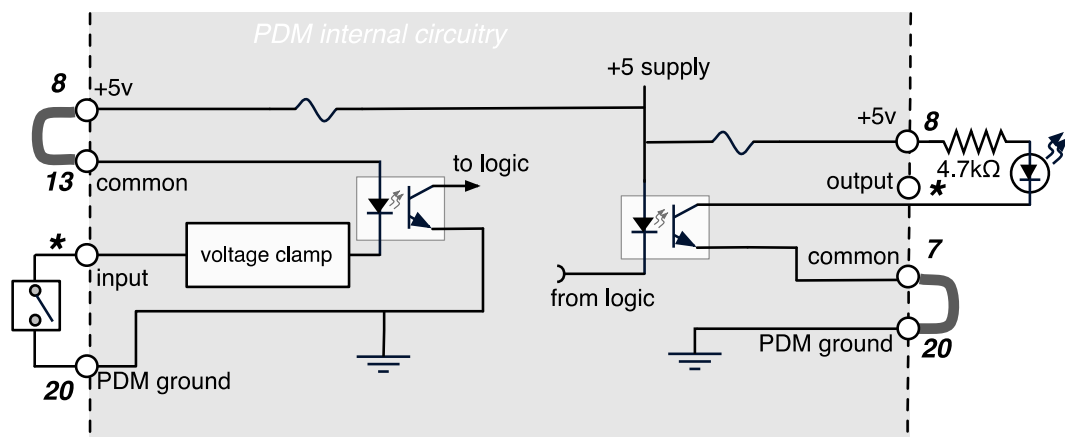
Eight parallel control inputs and eight parallel control outputs appear on the female DB-25 connector, with pinouts detailed below. Input and output functions are assigned through the GPIO menu.

Inputs and outputs are opto-isolated for easy interface to other equipment. A +5v supply and ground are also brought out to the DB-25 for simple remote controls using pushbuttons and LED status readouts. If an output signal is a pulse, its opto-isolator will turn on for between 100ms and 1000ms (adjustable on the configuration screen). Factory default is 250ms.

The +5v supply can provide 200 mA, more than adequate for 8 LEDs and 8 logic inputs. Total draw should not exceed this value. This supply is protected by an internal, self-resetting thermal circuit breaker. If this breaker should trip, power off and allow several minutes for it to reset.

Basic remote control

1. Use the connections shown on the left side of the schematic below. At the DB-25, jump pin 13 (input common) to pin 8 (+5v).
2. Connect one side of an SPST NO pushbutton to pin 20 (ground); connect the other side to pin 25 (Input 1).
3. Add up to 7 more pushbuttons as needed, connecting one side of all to pin 13 as common, and connecting their other sides to the individual inputs in the chart below.



Typical PDM remote input (left) and output (right) circuits. Use the jumpers shown in gray for simple pushbutton / LED remote. Or use just the opto-isolator connections without pins 8 and 20 for interface to other equipment. DB-25 pinouts shown in bold. * indicates individual DB-25 pins for separate GPIO inputs and outputs; see chart below.

Basic remote readout

1. Use the connections shown on the right side of the schematic. At the DB-25, connect pin 7 (output common) to pin 20 (PDM ground).
2. Connect a 4.7 k Ω 1/4 watt resistor between pin 8 (+5 V) and the anode or + lead of an LED; connect the cathode or – lead of the LED to pin 19 (output 1).
3. Add up to 7 more resistor-and-LED combinations as needed, using pin 8 as the common voltage supply and connecting the LEDs cathodes to individual outputs in the chart below.

Connecting to external logic or relay circuits:

PDM's GPIO inputs and outputs are isolated from its power supply and ground, and can be connected to external logic circuits or the power supplies in remote equipment:

- Inputs require at least 6 mA flowing from the input common (pin 13) and the individual input circuit pin.
- Outputs can carry up to 25 mA, up to 30 V DC per channel, between the individual output circuit pin and output common (pin 7).

GPIO Pinouts

PIN	SIGNAL	PIN	SIGNAL
1	NC	14	NC
2	Ground	15	NC
3	Output 8	16	Output 7
4	Output 6	17	Output 5
5	Output 4	18	Output 3
6	Output 2	19	Output 1
7	Output Common	20	Ground
8	+5VDC	21	NC
9	Input 8	22	Input 7
10	Input 6	23	Input 5
11	Input 4	24	Input 3
12	Input 2	25	Input 1
13	Input Common	NC — No internal connection	

Serial Port

RS-232

The RS-232 pins are configured as a computer DTE port, similar to the Comm 1 jack on desktop computers.

If you want to control it from a computer or dumb terminal, you'll need a DB-25 null modem or a cable that reverses pins 2 and 3 (data), pins 4 and 5 (flow control). Since PDM uses standard pin configurations, you should be able to use an easily available DB-25 to DB-9 adapter to connect more common 9 pin cables.

RS-232 can be used either for control, or to delay data. In the case of delays, data gets sent on the same RS-232 connector as it's received, so if you have one device generating data and another one listening, you'll need a Y cable to split the TX and RX signals.

RS-485

There are also two independent RS-485 differential data lines on this connector. Each is independently controlled and programmable as input, output, or bi-directional.

Serial Pinouts on DB-25

PIN	SIGNAL	
1	NC	
2	RS-232 TX	Output from PDM
3	RS-232 RX	Input to PDM
4	RS-232 RTS	Do not use
5	RS-232 CTS	Do not use
6	No Internal Connection	
7	Ground	
8	RS-485 Port 1 +	Differential pair. Bi-directional.
9	RS-485 Port 1 -	
10	No Internal Connection	
11	RS-485 Port 2 +	Differential pair. Bi-directional.
12	RS-485 Port 2 -	
13 - 25	No Internal Connection	NC — No internal connection

Axia / Livewire setup

Introduction

PDM's optional Livewire version lets you connect your profanity delay system directly to a Livewire network. Once configured, you can:

- Feed PDM with any signal available on your system
- Send its full-quality output to any available input (or multiple inputs at the same time)
- Control PDM directly from selected Axia consoles
- access PDM's GPIO system directly within a console, or at standalone nodes anywhere on the network.

All this is carried over a single Ethernet cable, from PDM's back panel to the closest Livewire network switch. No additional PDM wiring is needed; in fact, Axia-equipped PDMs don't even have active analog or AES/EBU audio ports.

All controls, and most of the menus, are identical to PDM's front panel. Operators won't have to learn anything new. Installation and setup is almost the same, with a few important exceptions.

For that reason, read the rest of this manual first. Then use this section to understand Livewire-specific setups.

If you have an analog | AES/EBU version of PDM and wish to convert it to run as a Livewire device, please contact customer support. .

Network Setup

Since PDM-Livewire uses network connections instead of standard analog or digital audio wiring, it won't operate until it's been properly configured for your network.

PDM's output audio is broadcast to the network as a single **Livewire Source Channel**, identified by a number between 1 and 32,767. This number should be unique on your network, and not be assigned to any other source channel.

PDM's input is fed by specifying a **Livewire Destination Channel** number. This is the Source Channel number of some other device on the network. It's often the output of a console, but it can be a network or remote, or any other signal source.

PDM's Livewire GPIO system provides multiple sets of inputs and outputs, in four groups of five i/o each. Each group is assigned a unique address.

- Any of the 20 inputs can be assigned any Input function from PDM's GPIO list ([page 35](#)). The same function may be set to trigger from multiple inputs.
- Any of the 20 outputs can carry any Output state from the GPIO list. A state may be assigned to multiple outputs.

- Pairs of inputs and outputs don't have to be assigned from the same row on the GPIO list on page 27. For example, Input 2 can be assigned to the Cough Button while Output 2 monitors Cue 1. They also don't have to match assignments in any other group. You are free to assign them any way you want³¹.
- PDM's hardware GPIO is independent of Livewire. You can assign totally different functions from PDM's front panel (page 34).

If you are controlling your network with **Pathfinder**™ routing and control software, you have to register PDM in Pathfinder's configuration controls. If Pathfinder doesn't know PDM is on the network, you won't be able to use that software to route its audio or GPIO.

Axia Addresses

For Livewire to work, your PDM must have an IP address, and audio and GPIO channels have to be specified.

1. Press **Menu** and then **NET** to open the Network menu (page 32).
2. Axia uses static addresses for IP audio networks, so make sure **DHCP** is **Disabled**.
3. Manually set a valid **IP Address** and **Subnet Mask** according to the instructions on page 25. You may need to contact your network administrator or Axia instructions for specific information. Save the changes, and power cycle PDM so they get applied.

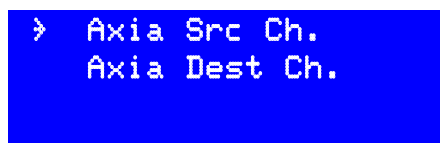
Once you've done this setup, view the **INFORMATION** page to verify PDM's IP address. Then you can check the connection by **Pinging**³² that address from another computer on your network.

If you want to connect PDM to both an Axia system and a distant NTP server (page 33), you'll need to custom-configure your router. Contact your system administrator.

Livewire Audio Channels

25-Seven Systems follows Livewire conventions for *source* and *destination*. These channels may only be set from PDM's front panel LCD interface.

Open the **Network** menu on a Livewire-equipped PDM, and select the **Axia** menu choice. You'll see a screen like this:



This screen lets you select Source or Destination channel menus.

Use **↕** or **↕** buttons to select which menu you want. Confirm your choice with **↵** or press **⊗** to cancel.

The input to your PDM must be some other audio stream on the network, such as a network feed or the main output of a talk studio console. We call PDM the **destination**³³ for this signal, identified by the Axia channel number of the device broadcasting it.

³¹ Axia has recommendations for using PDM's GPIO with Element consoles, to make the operator experience as seamless as possible. Contact them for instructions.

³² "Ping" is a network diagnostic tool provided with almost all computers.

³³ It may seem counter-intuitive to set up PDM's input with a "Destination" channel. We did this to be consistent with other Axia equipment.

Select `Axia Dest. Ch` from the `Axia` menu. The LCD will look like this:

```

Axia Dest.Channel:
Pgm2@ LivewireStudio
00120          OK✓
<→↑↓Chg value  CNCLX
  
```

This screen lets you set the Livewire Destination channel. As you scroll through channel numbers, this line shows the name of each signal available on the Axia network

Use `↕` or `↖` buttons to increment; `↩` or `➤` to jump by 100's.

Confirm your choice with `⬇` or press `⊗` to cancel.

- Remember: on the `Axia Dest` screen, you decide which signal this PDM is the **destination** for. Enter the channel number of **some other source** on the network. PDM will treat that source as its input.

Once you've set an address in this screen, PDM checks the network and makes sure the address is valid.

- If that address isn't valid, the front panel AUDIO LEDs remain amber.
- If the address is valid but the source isn't passing audio, the LEDs go dark.
- If there is a valid audio signal, the LEDs light green to indicate its level.
- You can also check the signal flow by putting PDM in real-time play, feeding it a signal, and listening to PDM's Source Address on any Axia control surface or other monitor point.

Your PDM's output must be designated a **source**³⁴ before other devices on the network can listen to it. Assign it a channel number in the `Axia Source` screen. Other devices will be able to hear PDM's output by using this channel number as an assigned Destination signal.

Select `Axia Src Ch.` from the `Axia` menu. The LCD will look like this:

```

Axia Source Channel:
Pgm1@ LivewireStudio
00220          OK✓
<→↑↓Chg value  CNCLX
  
```

This screen lets you set the Livewire Source channel.

As you scroll through channel numbers, this line shows the name of each device available on the Axia network

Use `↕` or `↖` buttons to change channel numbers by single digits.

Use `↩` or `➤` to jump by 100's.

Confirm your choice with `⬇` or press `⊗` to cancel.

³⁴ Again: we've followed Axia's standard for signal names, to keep setup as simple as possible.

Livewire GPIO

PDM-Axia units provide up to 20 inputs and 20 outputs of Livewire IP-based GPIO, with separate function choices for each. This is in addition to the parallel DB-25 GPIO supplied with all PDMs (page 34). The two interfaces can be programmed and used independently.

The Livewire GPIO is configured from PDM's Graphical User Interface, on a networked computer.

Log into the GUI (page 39) and click the **CONFIGURATION** tab. In addition to the functions described on page 44, Axia equipped PDMs have the options in the following screenshot.

The screenshot displays the 'Livewire GPIO' configuration window. It is titled 'Livewire GPIO' and has a 'Livewire GPIO: Enabled' checkbox. Below this, there are three port configuration sections:

- Port 1:** Port 1 Address: 259. Inputs: In 1 (Dump), In 2 (Exit), In 3 (Cough), In 4 (Build), In 5 (None). Outputs: Out 1 (DumpLamp), Out 2 (ExitLamp), Out 3 (CoughLamp), Out 4 (BuildLamp), Out 5 (DumpReady).
- Port 2:** Port 2 Address: 192.168.1.21/4. Inputs: In 1 (Dump), In 2 (Build), In 3 (Exit), In 4 (Cough), In 5 (Bypass). Outputs: Out 1 (DumpLamp), Out 2 (ExitLamp), Out 3 (DelayFull), Out 4 (DumpReady), Out 5 (None).
- Port 3:** Port 3 Address: (empty field).

Ports 3 and 4 (not shown) offer the same drop-down menus as Ports 1 and 2.

PDM provides four Livewire ports with five lines each of user-selected input and output control.

- Each port's section contains a Port Address entry line. Type the address of a Livewire device you want PDM to communicate with. Devices are typically Element consoles with an Expert Monitor module, or Axia General Purpose Input/Output Nodes.
- Input and Output selections for PDM Livewire GPIO are the same as for the PDM's Hardware GPIO. See page 35 for a complete list.

GPIO with Axia Element consoles

We have a recommended configuration for PDMs that are being used in Element studios. Use it, and PDM will follow the button labels already printed on the console.

Enter the channel number for that Element console. In the screen shot above, PDM Port Address 1 is set to talk with a console on channel 259.

The screen example here shows our recommended GPIO settings for working with an Element Delay section. They let operators control PDM by following the button labels already printed on the Element console.

In1=Dump	Out1=DumpLamp
In2=Exit	Out2=ExitLamp
In3=Cough	Out3=CoughLamp
In4=Build	Out4=BuildLamp
In5=None	Out5=DumpReady

You also have to configure the console itself, so messages are properly sent to PDM. You'll need a computer connected to the Livewire network, and a browser.

1. From the web browser, enter the IP address of this particular console. This will take you to the Element Control Center page.
2. In the left-hand column, under Sources & Profiles, select Configuration. A new page appears which contains a list of your Show Profiles.
3. Select your current Show Profile from the list. A page listing the Show Profile's configuration sections will appear.
4. Select Monitor section. A page appears where you can adjust monitor settings.
5. At the end of the General Monitor Settings section is a field labeled GPIO channel for external profanity delay. Enter the channel number the Element should use to send Delay GPIO messages to the PDM. In our example, it would be 259.
6. Scroll to the bottom of the form and click Save changes.
7. On the Element console itself, press the PROFILE button on the GLOBAL OPTIONS panel.
8. Select your PROFILE again to load the changes you just made into the console. Element will now send GPIO messages (over the channel selected in step 5), when you press any of the DELAY buttons on the CHANNEL OPTIONS panel.
9. Press PROFILE again to leave the CONFIGURATION screen.

Other Livewire GPIO

PDM can also receive triggers and send status levels to other distant devices, using an Axia GPIO node. You may want one of these near an automation system or satellite terminal. Set it up by opening PDM's GUI on a network computer, clicking the **CONFIGURATION** tab, and entering an address in one of the Livewire Port fields.

- The format is *IP address of the Node*, a slash, then the node's *port number*.

Axia GPIO nodes typically have eight separate DB-15 connectors. Each is electrically separate, and represents a different port on the unit. So the port number will be between 1 and 8, depending on which of the node's jacks you're using.

In our screen shot example, PDM's Port Address 2 is set to communicate with port 4 of the GPIO Node at 192.168.1.21... In other words, if you pull Pin 11 (GPIO input 1) low on DB-15 number 4 (Port 4) on that node, PDM goes into Dump mode.

Of course you're free to use any input or output, on any port of any node on the network, and assign any PDM function.



About the Bypass button

In standard XLR analog/digital versions of PDM, the Bypass button takes PDM out of the signal chain. Hardware relays connect PDM's analog and digital inputs to its outputs with no processing or delay. This design acts as a failsafe, even in the event of a power failure.

But in an IP based audio world, there is no equivalent. Relays can't route IP packets. If the power has failed, there's no way to route packets at all.

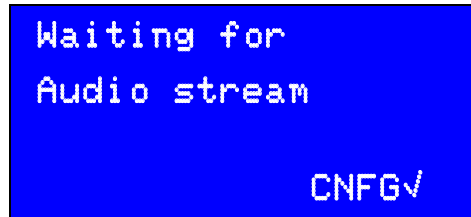
Therefore, pressing **BYPASS** on PDM-Axia units has the following effect:

- Any existing buffer gets dumped. BUILD, EXIT and other controls become inactive.
- Inbound packets get sent out, bypassing any PDM-specific functions. There is, however, a tiny delay while the packets pass through PDM's computer.

You may find it useful to set up Pathfinder software so that PDM's Bypass button actually changes the IP routing, and removes PDM from your Livewire air chain.

Livewire Troubleshooting

During startup, PDM's LCD will look like this for a moment:



This means that PDM is logging on to the Livewire network.

Use  to access the Configuration menu.

The screen will go away after PDM has successfully logged onto the Livewire network. Once it is gone, PDM's LED level meter should switch from yellow to green. You should see the meter bounce when IP audio is being properly sent to PDM.

If the `Audio stream` message remains up for several minutes after you start PDM, check the following:

- Does the INFO screen show a valid IP address?
 - Remember, Axia uses Static IP addresses assigned by the user, not DHCP addresses automatically provided by the router. If your PDM is set for DHCP, follow the instructions on [page 69](#).
- If the address is valid, can you Ping this address from another machine on the same network segment?
 - If not, check network connections, cables, and switches.
 - Are you using the correct Network Mask? Some Livewire networks may be configured on a class B Subnet. Try changing the Network Mask setting to 255.255.0.0
- Does your Pathfinder router software know PDM is on the network? You may need to add PDM as a listed device within Pathfinder.
- Is PDM as a destination (`Axia Dest`) set to a valid *source channel number* for some other device on your Livewire network? If it isn't, PDM's LEDs will be amber and the `Waiting for Audio Stream` message will remain.

Once you're sure audio is being sent to PDM, the unit will start in Real-Time Play mode. Now you should be able to monitor PDM at its Source address (as set in the `Axia Source` screen), from other devices on the Axia network. Verify meter and monitor levels to and from the PDM at your Livewire node or control surface.

Avoiding Trouble

PDM's link to the network should never be disconnected while the unit is in normal operation. Loss of network connection while the unit is operating will result in a loss of audio, and `Waiting for Audio stream` gets displayed on the LCD.

Once the network connection is restored, PDM will return to Bypass or Real-Time Play, and you can continue normal operations.

Software Updates and Factory Access

PDM includes features that aren't normally available to the user, for maintenance and software updates.

Updates

25-Seven Systems is constantly working to improve our products. Please check in with us for the latest software release information. Users running PDM version 1.5 or higher can self-install emailed versions of the latest firmware using a hidden, web-based utility from a networked computer. Specific instructions will accompany any software release package.

Make sure PDM is in Bypass mode before beginning any software update.

Remote factory access

We have also provided a unique way for our software engineers to perform remote diagnostics, if there are ever any problems. **You have to specifically set up Remote Access before we can start. This setup has to be done before each Remote Access session. We absolutely can't get into your PDM unless you let us.**

The set up procedure requires a specific series of steps, each time you want to enable it. It's virtually impossible to perform these steps accidentally.

If you think remote access may be appropriate, contact us by phone or e-mail. We'll set up a time, and give you instructions for enabling the session. You may need to open port 22 on your firewall to facilitate connections.

Audio Specifications

S/N = 84 dB with 10 dB headroom (94 dB dynamic range), A-weighted

THD @ 1 kHz = .025%; IMD (IHF) = .05%

Frequency response = ± 0.5 dB 20 Hz – 20 kHz

Measured in all modes, using analog connections; expect even better results with digital input and output.

Warranty

Telos Alliance Limited Warranty



For the latest Telos Alliance warranty, visit: telosalliance.com/warranty



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