Versatile Systems from Barix and BSW

Barix Encoder/Decoder/Codec Audio Systems

The Barix Instreamer 100 MP3 audio encoder is a versatile, network-enabled analog and digital audio-to-Ethernet converter for broadcast audio distribution. It converts audio from an analog or digital device into high-quality MP3 streams. The audio is encoded in real-time, and the generated audio stream can be distributed, via an IP-based network or the Internet, to one or more receivers or Shoutcast/Icecast-servers. It can be easily managed via a web browser interface using PCs, web pads, PDAs or other web-enabled devices.

With serial and Ethernet control APIs, open IP-standards, and the MP3 format, the device can also be integrated with other components, controlled by automation systems, or used with Barix’s Exstreamer 100 MP3 audio decoder as a flexible, cost-effective networked audio system. The Exstreamer 100 provides unbalanced RCA outputs, and a headphone jack, along with Ethernet connection and web-browser control. The Exstreamer 100 is a versatile network audio decoder for a variety of applications. It enables users to create flexible, cost-effective distributed audio systems using standard IP technology. It plays MP3, G.711 & PC audio files from PC or web server as well as legacy digital and analog sources (using the Barix Instreamer) and streams from sources like Shoutcast, Icecast or RTP servers (Instreamer). It provides a backlit LCD showing status and metadata information, unbalanced RCA outputs and a headphone jack, along with Ethernet connection and web-browser control. The Exstreamer 500 is a professional IP Audio codec featuring balanced analog audio interfaces and contact closures. A USB flash interface allows use of external memory to store audio files. The low power, reliable architecture makes the device an ideal choice for STL applications, professional audio endpoints (with IP or Ethersound protocol) and generic audio routing applications.

The Annuncicom 100 is a versatile IP Intercom and PA device, capable of full duplex (bidirectional) streaming of uncompressed audio formats, such as PCM, u-law, or a-law, or half-duplex encoding or decoding of MP3 streams. With its two contact closure inputs and built-in relay, the Annuncicom 100 provides the necessary I/O for push to talk and door opener functionality. A built-in speaker output and microphone amplifier allow direct connection of microphone and speaker. Line level inputs and outputs are also available.

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Barix MP3 Encoder or Decoder with Balanced Digital I/O

The Barix Exstreamer 1000 is a versatile network device that can function either as an audio encoder or decoder for a variety of high quality audio applications. The Barix Exstreamer1000 features AES/EBU in and out to provide a fully digital transmission path.

- Encodes or decodes audio in high quality
- AES/EBU and balanced stereo audio interfaces (inputs and outputs) via 9-pin connection
- Supports streaming (http, UDP, RTP) with automatic failover and USB playback
- 10/100 Mbit Ethernet connection
- Contact closure interfaces (4 in, 4 relays)
- Control via standard web browser as well as serial, TCP, UDP, cgi API

Contact BSW For Lowest Prices

This just in: BSW’s fully stocked Ohio warehouse is dedicated to getting your order out the same day you place it!
I Great for STL!

Flexible Dual-Channel ISDN Codec

The EAGLE – a tried and true and well known dual-channel ISDN audio codec. Imagine having two remote broadcasts feeding audio to the studio, and all three locations communicating with one another! Eagle’s unique multiplexing capabilities does that and a whole lot more. The Eagle can also interface with many popular codecs on the market. And when ISDN lines aren’t readily available, EAGLE has a back-up POTS mode with frequency extender.

- Analog and Digital inputs and outputs
- Dual-channel for 64 or 128 kbps encoding modes
- G.711, G.722 and MPEG
- Mic input; headphone and front phone jacks
- VU metering of send/receive audio
- Built-in terminal adaptor for U.S.A. (Nat 1) and Europe (Euro ISDN)
- Silent operation (no fan)

Can’t decide? See our comparison chart p. 53

Cost-effective Audio Over IP Software Bundle

Audio Compass is a great software solution for your remote broadcasting needs. This special bundle gives you both software solutions: Audio Compass Navigator, for server based (GUI) audio over IP, and Audio Compass P2P, a point-to-point command-based broadcast option. With either software, after your initial setup, you are on the air with a single click.

AUDIOPASS P2P FEATURES:
- Bi-directional text chat
- G.722 Codec
- Automatic Gain Control
- Control over packet group size for those difficult wireless connections
- Improved audio quality and stability when other applications are running

AUDIOPASS NAVIGATOR FEATURES:
- See all broadcast locations in your account group
- Text-chat to all members of your team, instantly
- Nat Traversal Server
- UDP and TCP support
- Support for LAN, WAN, Public Internet and 3G Wireless Networks
- Remotely control audio switchers and send relay closures via RS232 serial port
- Direct control over its audio buffer
- Choose from the SPEEX, ILBC, G.711, G.722 and GSM Encoders
- Automatic Gain Control
- Industrial High-Speed Ethernet
- adaptive buffer mitigates network jitter. Built-in web server for IP based remote control and configuration. Also great as a totally flexible and complete ISDN solution. AEQ’s “PHOENIX POCKET”, “PHOENIX LITE”, and “PHOENIX PC” smart phone apps and associated smart phone microphone/ headphone cable kit(s) are great additions to use for connecting to this codec via 3G/4G/WiFi and your computer!

Codec and Mixer Platform

The PHOENIX MOBILE is an all-in-one portable codec, digital mixer, and communications platform. Has 2 mic inputs, 2 headphone outputs, 1 mic/line input, 1 stereo line level input, 1 line level stereo output, built-in ISDN, PSTN, and PSTN modules, protective cover and 12V DC power supply/charger. It features dual channels for backup and coordination/talkback, as well as an easy to navigate advanced user interface. Totally connection interoperability with the PHOENIX STUDIO or any N/ACIP compliant codec. Two slots for optional communication modules (POTS and ISDN). Great for doing remotes. AEQ’s new “PHOENIX POCKET”, “PHOENIX LITE”, and “PHOENIX PC” smart phone apps and associated smart phone microphone/headphone cable kit(s) are great additions to use for connecting to this codec via 3G/4G/WiFi and your computer!
THIS CODEC HAS BEEN THROUGH TWO WARS, MULTIPLE ELECTIONS, Floods, famine, earthquakes, hurricanes, marathons, championship games, regattas, LOTS OF CONCERTS AND WAY TOO MANY CLUB EVENTS TO ADMIT.

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BSW Has ACCESS Portable Accessories:

Be prepared for any potential SNAFU (no matter how remote) with these handy accessories. Cover all your power needs with a standalone battery charger, spare battery and 12V DC adaptor. And protect your Portable with a super-durable hard case (that can withstand the most ornery baggage handler) as well as soft foam inserts to hold and cushion it.

- ACCESSCHARGE: Standalone battery charger
- ACCESSBAT: Spare battery
- ACCESS12VDC: 12V DC Adaptor
- ACCESS-SMALLROADCASE: Hard travel case

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Comrex BRIC-Traversal Server Software

The BRIC Traversal Server (BRIC TS) makes connecting your ACCESS codecs fast and simple. Maintained by Comrex, it helps keep track of all of your ACCESS codecs no matter where they are on the public Internet. Simply log on to the BRIC TS server, create an account, define a group and list all of your ACCESS units that belong to that group. After each ACCESS is “BRIC TS enabled” and connected to the Internet, they all sync up. A "buddy list" of all of the codecs in your group will appear on the Connections section of each ACCESS showing connection status, IP address and availability. Just select the device that you want to connect to and hit the "Connect" button! Even if you are behind a firewall or on a private LAN, BRIC TS gets you connected. Call BSW today.

- BRIC-TS1: One license
- BRIC-TS5: Five licenses
- BRIC-TS10: Ten licenses

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ACCESS, Our Most Popular Field Codec

Now with two USB ports and support for select 4G Wireless cellular data service modems, the Comrex ACCESS 2USB provides even more flexibility to meet the demands of on-the-go broadcasters. ACCESS codecs now ship with AAC Coding Algorithms included as well as BRUTE (BRIC UDP Transmission Enhancement) to give you a complete complement of tools to get your remote broadcast on the air.

- ACCESS-2USB: Portable IP codec
- BRIC-TS1: BRIC Traversal Server for Access

Accessories:
- ACCESS-SMALLROADCASE: Small Case

Contact BSW For Lowest Prices

ACCESS Portable Mixing Module Adds Even Greater Versatility

This mixer/headphone management module turns the ACCESS Portable into a full-featured 5-channel stereo mixer/codec.

ACCESSMIXER: List $1,200.00

Contact BSW For Lowest Prices

ACCESS for Your Studio

The 1U ACCESS Rack is optimized for use in a studio or rack room, although it’s perfectly at home in a traveling rack. You can control it using a standard web browser and can access it from virtually anywhere.

ACCESSRACK: ACCESS Rack

List $3,000.00

Contact BSW For Lowest Prices

Comrex BRIC-Link. The Perfect Low Cost STL Solution!

The BRIC-Link is a high-quality, low-cost Audio Over IP solution for full duplex audio transmission over IP links. Simple to configure via a standard web browser, BRIC-Link can be used on a wide variety of IP circuits including T1, WAN, LAN, ISM band IP radios and satellite data links to work as a full blown codec OR as a great, low cost STL solution.

- BRIC-LINK: BRIC-Link Module

List $1,500.00

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Broadband Internet is everywhere. Which makes it ideal for live remotes.

Unfortunately, the internet is also notoriously erratic. Even if you’re lucky enough to get a good connection, it might deteriorate during your show. So you dial back the bitrate, sacrificing sound quality to play it safe.

There is a better way.

Meet the amazing new Z/IP ONE codec.

Telos and Fraunhofer collaborated to develop a unique coding algorithm that adapts to changing network conditions on-the-fly.

If your connection quality drops, the Z/IP ONE uses error concealment, elastic buffers, and bitrate adjustments to keep audio flowing at maximum quality.

The Z/IP ONE now makes it possible to use the internet for great sounding remotes.

The convenience of the Internet. The sound of Telos.

Contact BSW for The Best Price on all Telos products  800-426-8434
Telos: The Leader In IP/ISDN Codecs

The Best Sound For Low-Bitrate Audio
Zephyr Xstream codecs are specially designed for use with a single ISDN line or other low-bitrate transmission paths including IP networks.

- 2-channel flexibility over a single ISDN circuit (or two synchronous links)
- Can transmit and receive 20kHz stereo audio and from a single location, or 2 mono channels to and from separate locations
- IP streaming with both MP3 and AAC
- Available in 3 versions: rackmount, portable, with or without mixer

XPORT FEATURES:
- Custom DSP extracts maximum bit rate from standard phone lines
- Groundbreaking CT-accPlus coding improves efficiency 30% over MPEG AA
- Built-in web server for convenient remote and local control
- Independent headphone, receive audio and monitor mixes
- Ethernet port and DB-9 computer connections

XPORT-PI FEATURES:
- All XPORT POTS features as above, plus additional ISDN mode
- Low Delay AAC, equivalent to MPEG Layer III but with 75% less delay
- Supports G.722 for compatibility to other brand ISDN codecs

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IP Codec Handles Toughest Network Conditions
The Telos Z/IP ONE 1RU rackmount codec lets you use public IP networks and mobile phone data services for high quality, trouble free audio transmission with minimal setup and virtually no delay. It utilizes ACT (Agile Connection Technology) to continuously adapt to network conditions, minimizing the effects of packet loss, varying bandwidth, and jitter that can occur on real-world networks. With ACT, the Z/IP uses every bit of the available bandwidth for the best audio performance.

Agile Connection Technology (ACT) for flawless transmission
- Virtually bulletproof connection traverses firewalls and NATs
- Agile Connection Technology
- Audio I/O via AES/EBU, analog (balanced XLR) and Axia Livewire IP-Audio interface
- Studio-grade 24-bit A/D and D/A converters
- Supports SIP, G.711, G.722, MPEG Layer 2
- Wireless-capable; Wi-Fi, EVDO, UMTS and more

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Portable POTS/ISDN Codecs
The Telos Zephyr Xport POTS field codec lets you send high-quality audio over standard phone lines to a Zephyr Xstream ISDN codec back in the studio. The Xport-PI version adds ISDN capability.

XPORT  List $3,195.00
XPORT-PI  List $3,695.00
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Cutting Edge Codecs From Tieline And BSW

Seamless Field Reporting
The Tieline Commander G3 Modular Studio Codec—the only codec in the world that connects via POTS, ISDN and IP now connects with the Report-IT iPhone app for even more remotes. It boasts wired IP Audio on board and accepts all optional Tieline modules. It features comprehensive front-panel hardware control including 4 x LED PPM meters that can be configured to show send/return or user selectable channel mix. It can also be remotely controlled via the Tieline PC GUI.

Economical Bridge-IT for Remotes, STLs and More
Tieline’s Bridge-IT is a low cost point-to-point or multi-point IP audio solution that is ideal for STL, Studio-to-Studio and remotes. With Bridge-IT you need no computer. Offering professional grade audio connectors, it comes in a basic version or a pro version (offering AAC algorithms and multiple unicast).

- Full hardware front panel interface including navigation, LCD display, PPM metering and dialing keypad
- Web GUI for programming codec functionality and RS232 data connections
- G.711 G.722, MPEG Layer 2, plus low-delay Tieline Music and MusicPLUS algorithms
- Multiple-unicast over LANs and send 6 stereo connections, one with bidirectional audio (Pro version only)
- LC-AAC, HE-AAC v.1 and HE-AAC v.2 algorithms (Pro version only)

Give Your Tieline G3 Codec New Life!
Tieline’s 4G USB Module lets your codec interface with the Pantech UML290 4G LTE USB Modem (not included) and broadcast live over the Verizon Wireless™ 4G network (3G also supported). Compatible with Tieline’s Commander G3 Field and i-Mix G3 remote broadcast codecs, it supports both 4G LTE (700MHz) and CDMA 1xEV-DO Rev. A/Rev. B networks (800/1900 MHz), as per modem specifications.

Customize Your Codec!
The Commander G3 Field IP codec features two module slots that accept your choice of hardware plug-in modules. A single wired IP module is included.

- 2 expansion slots that accept your choice of IP, 3G/3.5G/4G, POTS/PSTN, ISDN, X.21, GSM and B-GAN satellite plug-in modules
- A suite of algorithms to suit all your live broadcasting requirements
- Configurable 14 x 8 cross-point Digital Matrix Audio Router (DMR)

Connect Your Mic to iPhone for Superior Audio
Tieline’s Mic Adapter is a 20Hz-20kHz CD-quality digital audio input device that enables you to attach your favorite dynamic professional microphone to an iPhone 4®. Other interfaces roll off quickly after 200Hz reducing the richness of the audio quality. Mic Adapter delivers a full octave of richer, warmer, low frequency response from 20Hz to 200Hz and perfectly reproduces the mid and high frequency ranges.

Give Your Tieline G3 Codec New Life!

Tieline’s 4G USB Module lets your codec interface with the Pantech UML290 4G LTE USB Modem (not included) and broadcast live over the Verizon Wireless™ 4G network (3G also supported). Compatible with Tieline’s Commander G3 Field and i-Mix G3 remote broadcast codecs, it supports both 4G LTE (700MHz) and CDMA 1xEV-DO Rev. A/Rev. B networks (800/1900 MHz), as per modem specifications.

USB-4G-MODULE

Connect Your Mic to iPhone for Superior Audio

Tieline’s Mic Adapter is a 20Hz-20kHz CD-quality digital audio input device that enables you to attach your favorite dynamic professional microphone to an iPhone 4®. Other interfaces roll off quickly after 200Hz reducing the richness of the audio quality. Mic Adapter delivers a full octave of richer, warmer, low frequency response from 20Hz to 200Hz and perfectly reproduces the mid and high frequency ranges.

MIC-ADAPTER

1-800-426-8434 • www.bswusa.com • info@bswusa.com
Portable and Studio
Codes w/Digital Recorder

Need a complete digital recorder and IP codec in one? The economical MAYAH Flashman II is all things in one chassis. This ultra-small and ergonomic device combines linear or MPEG audio recording with up to 20 kHz on-air bandwidth via IP or 3G (3G card sold separately). Its Automatic Jitter Compensation makes sure your quality is the best getting through your networks. It features 2 switchable Mic/Line XLR inputs with 48V phantom power combined with 1 stereo line out and 1 send/receive controllable stereo headphone output. The Flashman II provides MPEG L2/L3/AAC, MPEG 4 AAC HE, MPEG 4 ELD, G.711/G.722, and linear audio. You record simultaneously on SD-cards or USB sticks while being On Air! Or you record first and then send a digital audio file. This is the future of portable digital devices. Sporty can then bring your audio on-air via IP, ISDN, POTS and 3G (so you are ready broadband and POTS (cards sold separately) and is even FlashCast compatible with party vendors, and can talk to RTP and SIP, the basic technology for corporate and public networks. It offers G.711, G.722, MPEG L2/L3, MPEG-2/4 AAC, MPEG-4 HE AACv2, Linear Audio (16 - 24 bit), DAT (Lin.12 bit) algorithms with balanced XLR and AES/EBU I/O. The new ultra-economic C1131N has the same extensive feature set as its bigger brother, but comes with G.711, G.722, L2, L3 and Linear algorithms and can be user-upgraded with 48V phantom power combined with all other IP/ISDN codecs on the studio end. Reporting from any remote events, Sporty can then bring your audio on-air via IP, ISDN, POTS and 3G (so you are ready no matter what the situation). Sporty adds low delay with MPEG 4 AAC ELD. The larger MAYAH Sporty portable IP/ISDN codec provides 20 kHz bandwidth over 3G broadband and POTS (cards sold separately) and is even FlashCast compatible with many other IP/ISDN codecs on the studio end. Reporting from any remote events, Sporty can then bring your audio on-air via IP, ISDN, POTS and 3G (so you are ready no matter what the situation). Sporty adds low delay with MPEG 4 AAC ELD.

German-Engineered IP/ISDN Codes

MAYAH: the #1 codecs from Germany are always available at BSW! The C11 series offers many different models for IP and ISDN, including some with redundant IP connections for the most reliable streaming solution. Call your BSW sales representative today to discuss your MAYAH needs and get the right model for your application.

The C1131 (standard version) is a powerful, fully scalable bi-directional codec with display and keypad control that boasts the most modern audio via IP architecture and audio coding formats while providing an efficient economic solution for broadcasters. It offers G.711, G.722, L2, L3, MPEG 2/4 ACC, MPEG 4 AAC HEv2, MPEG 4 AAC ELD, Linear Audio (16-24 bit) and DAT algorithms with balanced XLR and AES/EBU I/O. The new ultra-economic C1131N has the same extensive feature set as its bigger brother, but comes with G.711, G.722, L2, L3 and Linear algorithms and can be user-upgraded with 48V phantom power combined with all other IP/ISDN codecs if needed.

The bi-directional C1141 is designed specifically for ISDN and IP, as an ISDN codec with IP Control for management or as a codec for both ISDN and IP. The ISDN stack boasts many d-channel protocols and international connectivity. It can connect to portable devices such as the MAYAH Sporty and Flashman II as well as other, third party vendors, and can talk to RTP and SIP, the basic technology for corporate and public networks. It offers G.711, G.722, MPEG L2/L3, MPEG-2/4 AAC, MPEG-4 HE AACv2, Linear (16 - 24 bit), DAT (Lin.12 bit) algorithms with balanced XLR and AES/EBU I/O and extensive LED metering.

Need a complete digital recorder and IP codec in one? The economical Flashman II is all things in one chassis. This ultra-small and ergonomic device combines linear or MPEG audio recording with up to 20 kHz on-air bandwidth via IP or 3G (3G card sold separately). Its Automatic Jitter Compensation makes sure your quality is the best getting through your networks. It features 2 switchable Mic/Line XLR inputs with 48V phantom power combined with 1 stereo line out and 1 send/receive controllable stereo headphone output. The Flashman II provides MPEG L2/L3/AAC, MPEG 4 AAC HE, MPEG 4 ELD, G.711/G.722, and linear audio. You record simultaneously on SD-cards or USB sticks while being On Air! Or you record first and then send a digital audio file. This is the future of portable digital devices.

The Sporty II is all things in one chassis. This ultra-small and ergonomic device combines linear or MPEG audio recording with up to 20 kHz on-air bandwidth via IP or 3G (3G card sold separately). Its Automatic Jitter Compensation makes sure your quality is the best getting through your networks. It features 2 switchable Mic/Line XLR inputs with 48V phantom power combined with 1 stereo line out and 1 send/receive controllable stereo headphone output. The Flashman II provides MPEG L2/L3/AAC, MPEG 4 AAC HE, MPEG 4 ELD, G.711/G.722, and linear audio. You record simultaneously on SD-cards or USB sticks while being On Air! Or you record first and then send a digital audio file. This is the future of portable digital devices.

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The top-of-the-line C1191 dual IP codec is what you need for 24/7 streaming audio via IP. The C1191 provides two fully redundant working Ethernet connections for ultra-rock solid IP transmissions. It utilizes FEC and AJC minimizing the effects of changing bandwidth and packet loss. It combines automatic strategies for concealment and audio format adaptation. With its user friendly browser web-interface to be accessed from Windows, MAC OS and Linux, it simplifies any complicated situation. Other features include: Audio I/O via AES/EBU and analog, headphone output; SD card slot for audio storage card; 24-bit A/D-D/A; web-server for browser control; optional apt-X and Eapt-X (sold separately); and more. See the entire C11 line at www.bswusa.com.

Try MAYAH and join its worldwide club of satisfied customers, and take advantage of our extra low pricing and budget-saving package deals!

"Can't decide? Call me for additional product info and detailed advice."
— Justin Warbreck, Sales Representative
WorldCast From APT

Fully Duplex Stereo Audio Codec
The APT WorldCast Equinox is a multi-algorithm, fully duplex, stereo audio codec offering IP, ISDN and X.21 / V.35 connections. Designed primarily for studio to transmitter links and inter-studio networking applications, it provides an ultra-reliable platform for the transport of broadcast-grade audio. It comes with APT's Codec Management Software that can provide network control from a single seat.

- Dual IP ports for separate streaming and control
- Multiple algorithms for maximum compatibility
- Redundant power supplies for maximum reliability
- Advanced facility remote control at studio and transmitter site

SureStream technology enables broadcast-grade audio over inexpensive IP links
- Dual IP ports for separate streaming applications or separate control/streaming
- DHCP for automatic configuration of IP connection settings
- Wide range of algorithms: Enhanced apt-X algorithm, Linear PCM, G.722, G.711, MPEG 1/2 Layer 2, MPEG Layer 3 CBR and MPEG Layer 3 VBR
- Non-destructive, cascade-resilient coding with Enhanced apt-X
- Embedded auxiliary data for transmission of RBDS / RDS or PAD
- Up to 4 opto-coupled inputs and up to 4 relay outputs
- Audio backup from SD card or USB
- Fast boot time for mission-critical environments
- Support for SNMP, Alarm and Event Logging

Affordable, Effective IP Audio Encoder and Decoder
The APT Stream-In Silver is a highly affordable IP audio encoder offering reliable and robust encoding of audio content over IP networks. A large range of algorithms including Enhanced apt-X, MPEG 4 AAC-HE and linear audio are available on the compact unit. It is fully compatible with all other APT and N/ACIP compliant codecs. The Stream-Out Silver IP Audio decoder handles a wide range of professional and consumer audio formats. It offers headphone monitoring, audio backup with either an on-board USB port or ShoutCast server, embedded auxiliary data and auto-detection of an incoming IP stream.

- Affordable, Effective IP Audio Encoder and Decoder
- Non-destructive, cascade-resilient coding with Enhanced apt-X
- Embedded auxiliary data for transmission of RBDS / RDS or PAD
- Up to 4 opto-coupled inputs and up to 4 relay outputs
- Audio backup from SD card or USB
- Fast boot time for mission-critical environments
- Support for SNMP, Alarm and Event Logging

Versatile, Modular IP Audio Codec
The APT WorldCast Astral IP audio codec is a rock-solid IP STL platform offering a full complement of professional audio delivery algorithms, including Linear PCM (16 bit and 24 bit resolution), Enhanced apt-X algorithm (16 bit and 24 bit), MPEG Layer II, MPEG Layer III, G.711 and G.722. With its modular architecture, broadcasters can then bolt-on additional functionality to meet their broadcast needs. Note: image shows Astral base unit with optional front panel.

- High quality audio transport over IP networks using RTP/UDP and SIP/SDP
- Dual IP ports for separate streaming and control
- Multiple algorithms for maximum compatibility
- Redundant power supplies for maximum reliability
- Advanced facility remote control at studio and transmitter site

Broadcast-Grade Audio Even Over Inexpensive IP
The APT WorldCast Horizon NextGen IP audio codec boasts patented SureStream technology to deliver the audio quality and reliability you expect from a T1/E1 link using inexpensive public Internet connections such as ADSL, wireless 3G and 4G, LAN, WAN and Wi-Fi. It also offers a wide range of algorithms as standard on a solid DSP platform with dual IP ports and redundant power supplies. Perfect for STLs and mission-critical applications, the WorldCast Horizon NextGen provides extensive control and monitoring capabilities to manage both your audio, data and network conditions and other equipment located at the transmitter site. Note: image shows base unit with optional front panel.

Why don’t we print our prices? We are constantly negotiating with our vendors, buying in bulk and putting together special packages to bring you the best price every day.
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<td>PHOENIXMOBILE</td>
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<td>3 ch mxr</td>
<td>Mono/Stereo</td>
<td>Multiple</td>
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<td>COMBASEFIELDIOP</td>
<td>IP + opt. POTS, ISDN</td>
<td>2 ch mxr</td>
<td>Mono/Stereo</td>
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