

UAPg2

Universal Audio Processor



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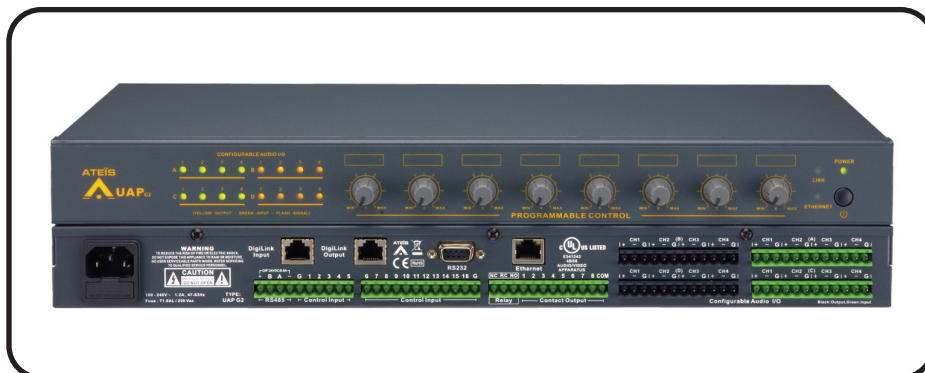
FEATURES

- Up to 16 analogue inputs/outputs
- Configuration via PC/Laptop
- Programmable, scalable front panel knobs for DSP functions
- Third Party control via RS232 or IP commands
- Full DSP drag and drop component library
- Up to 53 minute built-in message storage
- 16 built-in TTL inputs and 8 logic outputs
- Stack up to 12 units
- CE and UL listed
- 5 year warranty

DESCRIPTION

Designed for small to medium installations, the UAPg2 is a highly versatile drag n drop DSP processor. Hardware systems configuration is highly flexible in blocks of 2 for 4 in/12 out 6 in/ 10 out, etc. configurations. Expandability is also ensured by stacking multiple processors for a very high number of inputs and outputs. Front panel knobs can be programmed to map to any functions within the DSP library and can be scaled to match the requirements of the project. Up to 53 minutes of audio storage is built-in to the box and audio files can be activated via a built-in scheduler, any of 16 resident TTL inputs, or via an end user GUI interface custom built for each application. System design is PC based.

Software system design is PC based. Once connected the UAPg2 software, allows for design downloading, reversal of in-box designs, and live monitoring, calibration, an routing via the PC. Third Party control systems such as Crestron, AMX, Vity and others can also control the UAPg2 either via RS232 or IP commands. Up to 12 UAPg2 units can be networked together for expandability. A variety of accessories, including low cost analogue controllers (RAC), digital controllers (URC), and various paging and control microphones (PPM, PPM Touch) are available for use with the system.



SPECIFICATIONS

FREQUENCY RESPONSE: 20Hz - 20kHz @ +4dBu (± 0.6 dB)
DYNAMIC RANGE: > 105 dB
MAXIMUM GAIN: 66 dB
CROSSTALK: Line <-78dB, mic <-73dB
OUTPUT IMPEDANCE: 200 ohms
INPUT IMPEDANCE: 8k ohms
MAXIMUM OUTPUT: + 24dBu
MAXIMUM INPUT: + 24dBu
PHANTOM POWER: + 48 VDC
INPUT GAIN RANGE: 0 to 54 dB
SAMPLING RATE: 48kHz or 96 kHz (selectable)
A/D D/A CONVERTERS: 24 bit
POWER CONSUMPTION: <145 Watts
DIMENSIONS: 17.125"W x 1.75"H x 11.625"D
MAX WEIGHT: 8 lbs.
COMPLIANCE: CE LVD and EMC Directive, EU Directive 2002/95/EC, UL approved

SOFTWARE COMPONENTS:

- Delays – 5 ms to 2000 ms
- Dynamics – AGC (mono and stereo), Automatic Noise Sensing, Compressor, Comp-limiter, Expander (mono and stereo), Ducker (mono and stereo), Mono and Stereo Gate, Voice Gate, Gate with Sidechain
- Equalizers – Mono and Stereo GEQ (1 Octave, 2/3 Octave, 1/3 Octave), Mono and Stereo PEQ (2, 4, 6, 8, 10, 16 bands)
- Feedback Cancellation – Dynamic Feedback cancellation 1/5, 1/10, 1/20, and 1/100 Octave with 4, 8, 12, or 16 bands
- Inverter
- Level Controls – 1x1, 4x4, 8x8, 16x16
- Local Echo Suppression Module
- Logic – AND, NOT, OR, NOR gates with Net Input/Output for network applications
- Meters – 1 CH, 4 CH, 8 CH, 16 CH Peak/RMS meters
- Message Repeater – Up to 53 minutes of audio can be stored inside the box and output 2 separate messages simultaneously to independent zones. Messages can be activated using TTL inputs or via the built-in Scheduler
- Mixers – Automixers, Automixers with Mix Minus, Matrix Mixers, Standard Mixers, and Room Combiner
- Noise Generator – White, Pink, Tone
- Page Control Module – For zone paging applications
- Selectors – 4x1, 5x1, 6x1, 7x1, 8x1, 16x1, 32x1 for use with Third Party control or ATEIS RAC, URC remote controllers
- Custom Components – Build your own program within the program and password protect it

HARDWARE COMPONENTS:

- Main Frame – 4 slots for input/output cards, 16 TTL/Analogue Inputs, 8 TTL Outputs, RS485, RS232 Port, Ethernet Port
- Input Card – 4 Channels with individual Bypass, Mute, Sensitivity Select, Phantom Power, Mute, RTO (route to output), VU Meter, Signal Present, Level Control and adjustable Overload Threshold
- Output Card – 4 Channels with individual Overload Indicator, Mute, Meter, Signal Present, Level Control, and adjustable Overload Threshold
- Input/Output Card (with 2 Inputs and 2 Outputs) - individual Overload Indicator, Mute, Meter, Signal Present, Level Control, and adjustable Overload Threshold

■ ARCHITECTS' AND ENGINEERS' SPECIFICATIONS

The Digital Audio Platform shall be available with up to 16 inputs and outputs. Inputs/outputs shall be specified in blocks of 4, up to a total of 16. Mic/line Input and line level output options shall be available. Inputs/outputs shall be analog, with internal 24-bit A/D & D/A converters operating at a selectable sample rate of 48 kHz or 96 kHz. All internal processing shall be digital (DSP). Electronically balanced inputs and outputs shall be provided on plug-in barrier-strip connectors. Inputs shall be individually programmable for either microphone or line level audio signal and have assignable 48V Phantom Power. System shall be expandable to up to 12 total DSP units via RJ-45 Digital Links using standard CAT-5 data cabling, up to 32 feet between any two units max. Digital expansion links shall share up to 16 channels of digital audio at 48kHz or 5 channels of digital audio at 96kHz sampling rate between multiple units.

Internal system software shall be true "Drag and Drop" configuration with separate control/monitor GUI via direct data connection or via LAN/WAN using web-browser TCP/IP protocol. Available system audio program components shall include (but not be limited to) various forms of: mixers, equalizers, filters, crossovers, dynamics/gain controls, feedback controller, routers, delays, remote controls, meters, noise/tone generators, and diagnostics. Ethernet communications shall be utilized for local or remote software control, configuration, and DSP distribution. After initial programming, systems may be controlled/monitored using either TCP/IP or RS-232 serial communication by third party control systems (such as AMX® and Crestron®), by PC computer, and/or by dedicated remote control devices. Software shall operate on a PC computer running Windows® XP Professional/Vista/Windows 7.

DSP shall include two types of presets – "Master" and "Sub-Preset." There shall be a minimum of 32 Master Presets, which will allow user to switch the unit between totally different pre-programmed designs. There shall also be a minimum of 16 Sub-Presets that allow user to switch to pre-programmed parameter settings within each Master Preset.

The DSP shall incorporate an audio message player that can store up to 36-minutes of 8bit audio or 18-minutes of 16bit audio. Audio messages can be played either manually via direct PC selection, external analog or TTL logic input, Microphone paging station programmable flex button, or automatically via internal event scheduler. Messages can be manually or automatically routed to any single or multiple zones/outputs. There shall be up to 128 schedules with up to 100 programmable scheduled events per schedule.

The DSP unit shall incorporate 16 logic inputs to trigger events or presets and a other logic functions and 8 logic outputs with a common rail contact for controlling external functions. The unit shall also be equipped with 8 front panel control knobs, which shall be totally programmable for any variable function in the design program. This function knob shall also be 100% scalable to limit amount of volume/control. Front panel shall also have multi-color LED for displaying input/output signal presence, routing, and clipping indication for each input and output.

DSP shall contain both TTL Logic and RS-485 serial control ports for communicating with a myriad of intelligent remote controls and microphone paging stations. Remote Intelligent controls shall be either selector switch with volume knob in a 5-position or 8-position style (RAC-5 or RAC-8), or an LCD window style with control wheel and selector buttons – (URC for monochrome RS-485) or (URC-200 for multi—color IP-based controller). There shall also be the capability for remote intelligent gooseneck paging stations which incorporate either programmable flex buttons with expansion button units, or touchscreen controllers. Multiple remote serial control devices can also be daisy-chained by installing optional PPM-WJB junction box.

DSP unit shall be 1RU high, 17" x 1.75" x 9.5" (L x W x D). DSP unit shall be UL/C-UL listed and factory warrantied for 5 years.

■ ACCESSORIES



RAC5 / RAC8
5 / 8 Steps remote controller
(RS485).



URC200
Programmable remote controller (TCP/IP)



PPM/PPM-8
Serial programmable microphone
paging station



URC
Programmable remote controller
(RS485).



NSM
Sensing microphone



PPM-IT-5
IP touchscreen microphone
paging station