



## XAP™ 400 Audio Conferencing System

The XAP™ 400 combines a highly advanced digital matrix mixer that features Distributed Echo Cancellation®, noise cancellation, and audio processing with a single-line digital hybrid and a 10W amplifier. This combination provides a complete teleconferencing system—all in a single rack space unit.

In addition to Distributed Echo Cancellation, which places an echo canceller on every mic input, the XAP 400 features noise cancellation, which reduces background noise such as that caused by fans and air conditioning. The XAP 400 also features four configurable filters on each of the four microphone inputs. The result is crystal clear, echo-free audio in even the most challenging environments.

Echo cancellation performance is also enhanced by new EC reference summing capabilities. The XAP 400 has four configurable virtual EC references which allow you to reference multiple signals without sacrificing an analog output.

The XAP 400 is well suited as a stand-alone unit in small to medium rooms or as an addition to a video conferencing system—providing expanded audio capabilities including remote participation by telephone. For larger applications, the XAP 400 features an enhanced expansion bus, with 12 bi-directional audio buses, allowing it to be networked with other XAP products. Eight XAP 400s or 800s can be networked.

The XAP 400 supports four mic/line level inputs, four line level inputs, eight line level outputs, and a speaker output. Using G-Ware™ configuration software, you can easily route any input to any output or combination of outputs in a fully configurable matrix.



**The XAP 400 features a built-in telephone interface for teleconferencing capabilities.**

### Features and Benefits

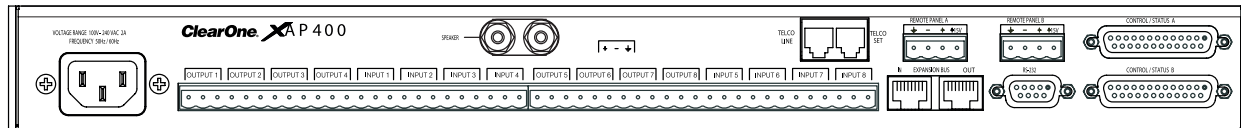
- Distributed Echo Cancellation
- Noise cancellation
- Four virtual references for echo cancellation summing
- Four configurable filters on each of the four microphone inputs
- Four independent processing blocks, each with 15 filters, delay and compressors, provide pinpoint audio configuration
- Front panel control of mute and gain on inputs and outputs
- Safety mute button on the toolbar to mute all outputs if feedback occurs during the configuration process
- 32 presets that can be executed without disturbing ongoing operations
- Up to 80 feet (24 meters) between connected units—eight XAP units can be linked for up to 64 microphones
- 10W, 4–16Ω speaker output
- Country-specific telco configuration settings
- Built-in telephone interface with telco noise cancellation and echo cancellation
- Control through expansion bus, RS-232 port, contact closure, ClearOne Control Panels, or XAP IR Remote

### Applications

- Conference Rooms
- Boardrooms
- Distance Learning
- Training Rooms
- Courtrooms
- Hotels

**The XAP 400 is manufactured and marketed by ClearOne, formerly Gentner.**

# audio conferencing



## Specifications

### Dimensions (LxDxH)

17.25" x 10.25" x 1.75"  
43.8 x 26 x 4.5 cm

### Weight

9.9 lb/4.5 kg dry  
13 lb/5.9 kg shipping

### Operating Temperature

32 to 100° F/0 to 38° C

### Humidity

15% to 80%, non-condensing

### Power Input Range

Auto-adjusting  
100–240VAC; 50/60Hz

### Power Consumption

30W typical

### Expansion Bus In/Out

Proprietary Network  
RJ-45 (2), 115.2kbps, 110k $\Omega$   
impedance  
Category five twisted-pair cable  
80' (24 meters) maximum cable  
length between any two XAP 800s,  
XAP 400s, or PSR1212s

### RS-232

DB-9 female  
9,600/19,200/38,400(default)/57,600  
baud rate; 8 bits, 1 stop, no parity  
Hardware flow control on  
(default)/off

### Control/Status

DB-25 female A/B (2)  
Inputs A/B: active low (pull to ground)  
Outputs A/B: open collector, 40VDC  
max, 40mA each  
+5VDC pins (2) (300mA over-current  
protected)

### Remote Panels A/B

4-pin push-on terminal block  
RS-485 Proprietary Protocol  
Category five twisted-pair cable  
1 pair data, 1 pair power and ground  
+15VDC (300mA over-current  
protected)

### Mic/Line Inputs 1-4

Push-on terminal block, balanced,  
bridging  
Impedance: 5k $\Omega$   
Nominal Level: adjustable -55dBu,  
-25dBu, 0dBu  
Maximum Level: -35dBu, -5dBu,  
+20dBu  
Echo Cancellation: 130ms tail time  
(works with 12dB of room gain)  
Noise Cancellation: 6–15dB  
attenuation  
Phantom Power: 24V, selectable

### Line Inputs 5-8

Push-on terminal block, balanced,  
bridging  
Impedance: >10k $\Omega$

Nominal Level: 0dBu  
Maximum Level: 20dBu

### Outputs 1-8

Push-on terminal block, balanced  
Impedance: 50 $\Omega$   
Nominal Level: 0dBu  
Maximum Level: 20dBu

### Audio Performance

Conditions: Unless otherwise specified,  
all measurements are performed with  
a 20Hz to 20kHz BW limit (no  
weighting)  
Frequency Response: 20Hz to 20kHz  
 $\pm$ 1dB  
Noise (EIN): -126dBu, 20kHz BW,  
max gain,  $R_s=150\Omega$   
THD+N: <0.02%  
SNR: 80dB re 0dBu (A-weighted)  
Dynamic Range: 100dB (A-weighted)  
Crosstalk <-91dB re 20dBu @  
20kHz channel to channel

### Assignable Processing Blocks

Filters:  
All pass  
Low pass  
High pass  
Low shelving  
High shelving  
Parametric EQ  
Notch  
CD Horn

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## Crossovers:

- Bessel
- Butterworth
- Linkwitz-Riley

Compressor; adjustable

Delay; adjustable up to 250ms

## Matrix Mixing Parameters

25x26 matrix

8 analog in/out

1 speaker out (10W)

12 expansion bus in/out

4 assignable processing blocks in/out

1 telco in/out

## Auto Mixer Parameters

Number of Open Microphones (NOM)

PA Adaptive Mode

First Mic Priority Mode

Last Mic Mode

Maximum # of Mics Mode

Ambient Level

Gate Threshold Adjust

Off Attenuation Adjust

Hold Time

Decay Rate

## Microphone Input Configuration

Input Gain Adjust

Mic or Line Level

Phantom Power on/off

Echo Cancellation on/off

Noise Cancellation on/off

Filters

All Pass

Low Pass

High Pass

Notch

PEQ

Mute on/off

Chairman Override on/off

AGC on/off

Speech Leveler on/off

Auto Gate/Manual gate

Adaptive Ambient on/off

## Telco Line

RJ-11

POTS (plain old telephone service) or analog extension from a PBX

A-lead supervision provided

## Telco Set

RJ-11

Connect analog telephone set

A-lead supervision provided

## Telephone Audio Performance

Conditions: Unless otherwise specified, all measurements are performed with AGC disabled, referenced @ -15dBm on/off the telephone line

Frequency Response: 250Hz to

3.3kHz  $\pm$ 1dB

THD+N: <0.2%, 250Hz to 3.3kHz

SNR: >62dB re max level

Pre-emphasis: 4dB @ 2kHz

## Telephone Echo Cancellation

Tail Time: 31ms

Null: >55dB

## Telephone Noise Cancellation

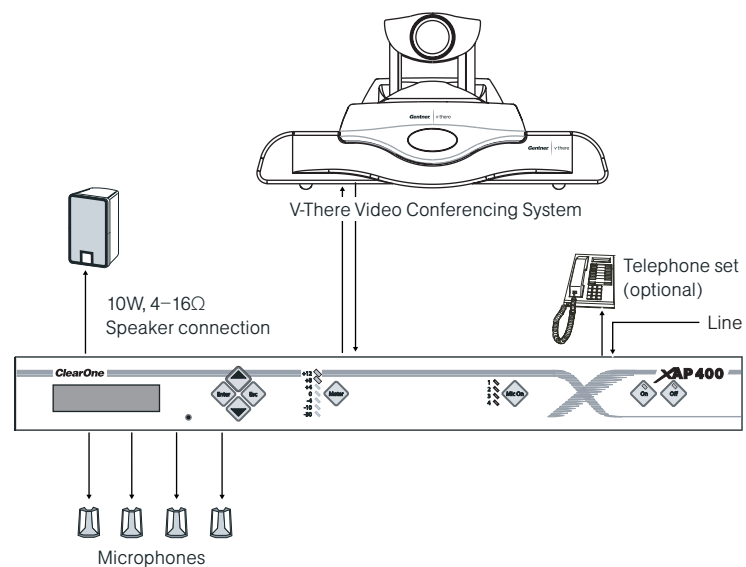
Noise Cancellation: 6–15dB attenuation

## Set-up Software

G-Ware

## Approvals

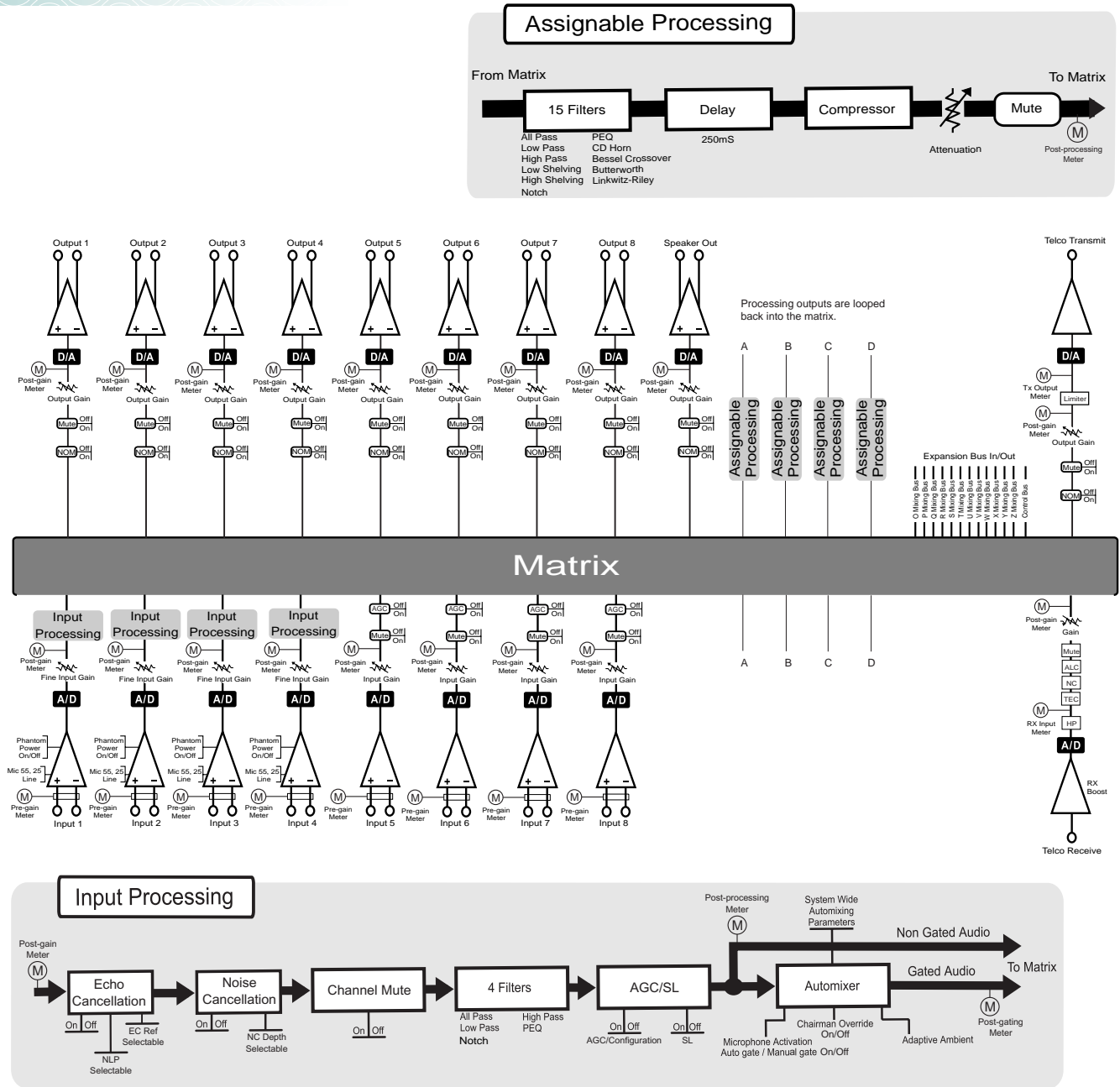
FCC, CSA, IC, CE, NOM, ACA, SABS, JATE



**XAP 400 connections in a typical video conferencing installation.**

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## Block Diagram



(M) = Meter Reference Point