



AudioMate 360a User ManualIP Voice Gateway for Audio Devices



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QUICK START

AM360 Configuration

www.voiceinterop.com

Register as a customer Choose the Support tab.

Download AudioMate 360 IP

Setup configuration utility

Utility

Quick Start Guide

Gather Your Tools

Web browser — Mozilla Firefox, Opera, Chrome, or Internet Explorer.

AM36Ø Configuration Utility — To find the address of each AM36Ø gateway, download the AudioMate 36Ø IP Setup utility from our web site, www.voiceinterop.com under the Support tab. You must first register as an owner of an AudioMate 36Ø gateway to be granted access.

(Optional) Telnet client — comes with Windows, or you can use a third-party client.

(Optional) USB Comm Port Driver — To view the boot and debug messages and set passwords you will need the USB comm port driver set from FTDI. These are quite common in many products so you might already have them installed.

(Optional) USB A to USB-mini B cable — To connect from your computer to the AM36Ø serial port

(Optional) Terminal program — HyperTerminal comes with Windows, or you can use a third-party datacomm terminal.

NO soldering iron nor screwdriver is needed to set up the gateway, just a web browser and a plan.

Basic Configuration Steps

- ¤ Find the IP address
- ¤ Loa in
- Set the RTP address and ports
- ¤ Connect the audio device
- x Set audio levels with a partner

Find the AudioMate 360 IP Address

Use the AM36Ø (AudioMate 36Ø) Configuration Utility to find the IP address that was assigned by your DHCP server. Click the **Locate** button and select a unit from the list on the left pane.

Log In to the AudioMate Unit with a Web Browser

Type the IP address of the desired unit from the list displayed by the AM36Ø Configuration Utility into the address bar of your web browser and press Enter. The AM36Ø will prompt you for the user name and password. The factory values are admin and password.

Set the RTP Address and Ports

The default AM360 settings might be all that you need to get going. If necessary, set the Remote RTP Address, RTP Remote Port, and RTP Local Port to communicate with your audio network.

Connect the Audio Device

Connect the audio device to the AudioMate 360a, whether it be a public address amplifier to the Line Output or a broadcast radio or an iPod to the Line Input. A microphone should be connected via the RJ45 on the front panel.





Set Audio Levels With a Partner

Set the Microphone, Line In, or Line Out Levels as appropriate to make the speech intelligble and pleasant to listen to. Be careful not to overdrive the AudioMate with too loud an input signal. This test requires two people: one to talk and another to listen.

See the Configuration section for detailed instructions.

To Set the Password with Telnet

To set the username and password on the AM36Ø use telnet to the IP address discovered by the AM36Ø Configuration Utility. The factory default username is admin and the password is password. After logging in, type the command 'help' to list the available commands.

To Set the Password via USB Serial Port

If telnet is not available, as a last resort the USB serial port can be used after installing the FTDI USB comm port emulators on your computer. These drivers can be downloaded from the FTDI web site. The comm port parameters are 115200 bps, 8 data bits, no parity, 1 stop bit. Once connected, send Ctrl-C (Control-C) to switch the unit to command mode. Type the command 'help' to list the available commands.

Please note that there are no audio settings available via telnet nor serial comm port, these must be configured via a web browser.

USB Drivers

The USB comm port drivers can be downloaded directly from the FTDI web site: ftdichip.com/FTDrivers.htm



TOUR

AudioMate 360a Product TourFront Panel

Audio	Nate 360a		
OK Link RX TX	Speaker Microphone Line Out Line In	Volume	Microphone
www.voic	einterop.com		

Front Panel Indicators

All front panel indicator LEDs can display one of seven color states: OFF, RED, RED FLASHING, GREEN, GREEN FLASHING, AMBER and AMBER FLASHING each of which can be used to indicate a different condition within the AudioMate 360a.

LED	Amber	Amber flashing	Green	Green flashing	Red	Red flashing
ОК	Booting, awaiting DHCP address	N/A	Normal operation	N/A	Boot error (view boot via USB serial port)	N/A
Link	N/A	Check LAN connection	LAN connection OK	AMCP data traffic received	N/A	N/A
Rx	Buffering audio from IP network	N/A	Playing RTP stream out audio port	N/A	N/A	N/A
Тх	N/A	N/A	N/A	N/A	Sending RTP stream to LAN	N/A
Speaker	Maximum volume	N/A	Mid-range	N/A	Minimum volume	N/A
Mic.	Maximum mic level	N/A	Mid-range	N/A	Minimum mic level	N/A
Line out	Maximum line level	N/A	Mid-range	N/A	Minimum line level	N/A
Line in	Maximum line level	N/A	Mid-range	N/A	Minimum line level	N/A
Pressing the Volume knob will select each audio line for adjustment.						



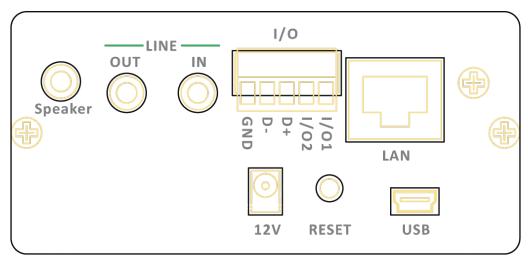
Front Panel Connections

Microphone — Accepts an 8-pin RJ-45 microphone plug on a Kenwoood KMC-27B pushto-talk mobile radio microphone. Other microphones might work and you may construct a suitable microphone using the pin-outs listed in Appendix A.

Front Panel Controls

Volume — The Volume control serves two functions: 1) selecting which audio line to adjust, and 2) adjusting the audio level. Select which of the four audio lines to adjust by pushing in the Volume knob and observing the indicator lamps on the front panel. The audio line whose indicator is illuminated will be adjusted by turning the Volume knob. There are 16 steps in the adjustment even though the Volume knob turns continuously. The lowest level is indicated by a red lamp, the highest level is indicated by an amber lamp, while all steps in between these extremes are indicated by a green lamp. The button presses and the level steps are echoed on the USB serial diagnostic port. Additionally, the resulting level settings may be observed on the unit's internal web page management screen.

Rear Panel



Rear Panel Indicators

LAN Port — The LAN jack has two built-in lamps: a solid green one indicating that an electrical link has been established, the other flashing green indicating packet activity.

Rear Panel Connections

Speaker — This 3.5mm mini phone jack allows connection of an 8-Ohm speaker to provide a simple means to get audio from the network into a small space such as a desk cubicle or small office. The audio power output is approximately 1/2 Watt. A monaural plug is preferred or use the tip of a stereo plug leaving the sleeve open.

Line Out — This 3.5mm mini phone jack allows connection to a line-level input of a public address amplifier, tape recorder, digital audio recorder, or other audio device. Monaural or stereo plugs will work properly in this port.

Line In — This 3.5mm mini phone jack allows connection of a line-level audio device, such as a CD player, iPod, tape recorder line output, AM/FM broadcast receiver, or NOAA weather radio receiver. Monaural or stereo plugs will work properly in this port.





I/O — Digital Input and Output lines. A 5-pin inline receptacle whose mating plug uses screw terminals to accept bare wire.

GND — Signal ground.

I/O 1 — (COR) Pull low to stream RTP audio packets to the network.

I/O 2 — (PTT) Goes low to indicate RTP packets are being received from the network.

D+ and D- — RS485 data loop; not implemented.

LAN — Standard 10/100BaseT Ethernet port with RJ-45 pin-outs following the AT&T T568B specification. A standard Category 5 LAN cable connects it to an Ethernet switch.

USB — Allows a trained technician with a correctly setup computer to establish a direct serial connection to the AudioMate 36Ø internal Diagnostic Console. Connects via a standard USB A to Mini-B cable. The computer must have the Virtual Comm. Port driver from FTDI (**ftdichip.com**) installed along with any standard terminal emulator. This port is used for diagnostics; no voice parameters can be set.

12 V — 12 Volt d.c. input, tip is positive, ring is negative. The supplied power adapter converts 120 Volt a.c. main power to 12 Volts d.c. at 500 mA maximum. The power plug has a 1.3mm inner diameter.

Rear Panel Controls

Reset — This recessed push button reboots the processor in the AudioMate 36Ø. Depress using a pen or any device with a small tip. The length of time that it is held depressed determines the type of reset:

Quick Reset – A momentary press performs a simple reset.

Factory Reset – Press and hold until OK light goes from AMBER to GREEN to reset those options found on the web configuration screens to factory defaults. While the AudioMate 360 is awaiting an address from a DHCP server this could take as much as 60 seconds. In order to ensure that the device loads factory default settings the Reset button must be held this entire time, until the OK light changes from AMBER to GREEN.

INSTALLATION

Installation

Network Considerations

In the typical AudioMate installation all endpoints are contained entirely in a corporate LAN, or participate via a WAN link, either unicast through a VPN or an engineered MPLS network, for example. It is preferable that all routers serving network segments that connect AudioMate endpoints be configured to route multicast streams. This can smooth implementation of multicast systems in general (such as videoconferencing), and the deployment of AudioMate Communicator remote control soft consoles in particular.

Multicast is used to allow multiple endpoints to share one stream of audio packets, thus reducing bandwidth consumption. However, while multicast streams typically pass seamlessly through Ethernet switches, they are typically blocked by most routers unless proper preparations are made to accommodate multicast. It is a near certainty that multicast





streams will not flow through most Internet routers. Therefore, unicast addresses are acceptable for use in the AudioMate 360 line of IP gateways, although multicast is preferred..

Bandwidth Consumption and Codecs

The G.711 codec (encoder/decoder algorithm) consumes approximately 84kilobits/second per stream. More aggressive codecs consume less bandwidth, but their output audio is typically of lower quality. Some codecs perform poorly in environments with high acoustic noise near the microphone. Narrowband IP links might require a more aggressive codec, while endpoints on your LAN will likely do well with the G.711 codec. Because the AudioMate Media Server can perform transcoding between algorithms you have the opportunity to optimize bandwidth usage versus audio quality.

Electrical Considerations

The AudioMate 360 gateways are powered by 12 Volts d.c. which can be provided by the supplied plug-in power supply or by a 12 Volt battery with float charging system. In either case, the power must be clean and free of a.c. noise components and interruptions. The supplied power module should be plugged into an uninterruptible power supply (UPS) for trouble-free service. If a 12 Volt battery/charger system is used, the charger must keep noise to less than 12mVp-p.

CONFIG

Configuration

Each unit requires two groups of configuration data: one group that is common across the AudioMate 360 family and a second group that is specific to the application of the unit.

Network Address

The units are shipped with DHCP enabled and will accept whatever address is assigned by your DHCP server. On startup the AM36Ø will wait up to 6Ø seconds for a DHCP server to assign its IP address. On timeout the AM36Ø will assign itself a default IP address of 192.168.0.254 and a netmask of 255.255.255.0.

AM360 Config Utility

The Address Finder utility is available by logging in to our web site:

www.voiceinterop.com Click the Support tab.

Locating the AM360 Address

Use the AudioMate 360 Configuration utility to locate the addresses of your units in preparation for configuration. Individual AudioMate 360 units are configured via web browser pointed to the unicast management IP address of the unit. The Address Finder utility is available by logging in to our web site

www.voiceinterop.com

Click the Support tab to find the link to download the AM36Ø Configuration Utility.

Open the AM36Ø Configuration Utility and click the **Locate** button and select a unit from the list on the left pane. The IP address of each unit is to the right of each minus sign.

Configuring by Web Browser

Type the address of the desired AM360 unit into the address bar of your web browser. (At the time of this writing the freely available Mozilla Firefox web browser provides smoother operation for these purposes.)





AudioMate 360 Login

username = admin password = password

Logging In

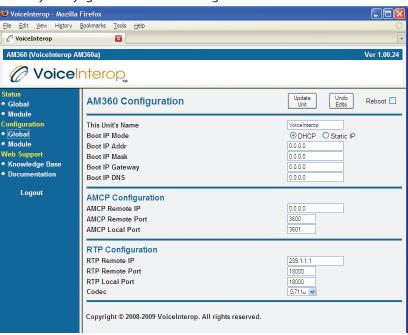
Like most network devices, the AM360 is protected by a password. The factory default for user name is admin and the factory default password is password.

When prompted by the web interface enter the appropriate user name and password. If they were accepted you will see a two part screen with a menu panel on the left pane and a page labeled AM360 Status on the right pane.

Global Configuration

In the menu pane on the left, locate the **Configuration** heading and click the link **Global** under it. The screen labeled **AM360 Configuration** should appear in the right pane.

You may safely ignore the **AMCP** settings.



This Unit's Name allows you to set a memorable name uniquely to identify this unit among the others on your network. This field accepts standard alphanumeric characters.

Boot IP Mode selects between automatic IP address assignment by your DHCP server or static IP address assignment specified in the fields that follow. If you select Static IP and then click the **Update Unit** button, the **IP Boot** fields will be filled with current values. If you select DHCP and click the **Update Unit** button the fields will be returned to Ø.Ø.Ø.Ø which indicates DHCP. Note that editing and saving the Boot IP fields has no effect on communications until you reboot the unit.

Boot IP Addr sets the static unicast management IP address used to configure this unit. Default is Ø.Ø.Ø.Ø for DHCP. Once you change this address, set the mask, update and reboot the unit, remember to point your web browser to the newly assigned address.

Boot IP Mask sets the netmask or network portion of the IP address. In a typical Class C network this will be 255.255.255.0 while the default value is 0.0.0 for DHCP use. If you set a static IP address you must also set this field to represent the size of your network.

Boot IP Addr

After changing the IP address of the AM36Ør you must reboot the unit to make the new IP address effective.

Also remember to change the address in your browser!

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Boot IP Gateway sets the default network gateway router that will carry your packets to other networks. If you set a static IP address you must also set the gateway to the default router on your network. The default value is Ø.Ø.Ø.Ø for DHCP use.

Boot IP DNS sets the IP address of your DNS server, if needed. Typically this address will be the same as your default gateway in small networks. Default is Ø.Ø.Ø.Ø for DHCP use.

AMCP Configuration

You may safely ignore the AMCP settings.

RTP Configuration

RTP Remote IP is the address to which audio packets will be sent. In multicast networks type the multicast group which will share audio with other units on the same multicast group. The Class D multicast range lies between 225.0.0.0 and 239.255.255.255 with numerous restrictions and considerations. Consult Appendix B for more information on multicast address assignment. For point-to-point unicast links type the unicast address (Boot IP Address) of the **other** end of the link, not this unit. A unicast address set here should fall within your network's private IP address space. Default is 239.1.1.1 multicast group.

RTP Remote Port is the IP port number assigned to this channel. This should be an even number between **16384** and **65534**. Default is 18000.

RTP Local Port should be identical to the RTP Remote Port number. Default is 18000.

Codec should be set to G711u for most applications. Other choices are G711a for G.711 A-law, and GSM for GSM 6.1 full rate. The G.711 codecs use about 84 kbits/second while the GSM codec uses about 13 kbs/sec. Default is G.711 µ-law codec.

Save Your Changes

Update Unit button — When you are satisfied with all the settings that appear on the **Global Configuration** page, be sure to click on the button labeled **Update Unit** at the top of the page. Your new settings take effect immediately, except for the fields labeled IP Boot. The IP Boot fields only take effect after a reboot.

Undo Edits button — To revert back to the previously saved values while editing click the button labeled **Undo Edits** at the top of the page.

Reboot checkbox — To reboot, check the **Reboot** box and click the **Update Unit** button. Any changes are saved before the reboot is executed.

Module Configuration

In the menu pane on the left, locate the **Configuration** heading and click the link **Module** under it. The screen labeled **AM360a** (**Audio**) **Config** should appear in the right pane.

Microphone Level displays the current setting of the Microphone input level. To set this level electronically, type in a value between Ø and 15 and click **Update Unit**.

Microphone Continuous Stream forces the AM36Øa to stream audio to the network continuously, regardless of the PTT input from the microphone. Check this box to force continuous streaming; clear this checkbox to require the PTT microphone digital input to control streaming to the network.

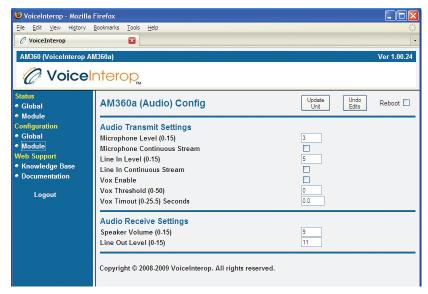
RTP Remote IP

is the address of other unit if using unicast, not the one that you are configuring.

Multicast Addressing

Appendix B has more detail on selecting multicast addresses with pointers to reference sources.





Line In Level displays the current setting of the Line Input level. To set this level electronically, type in a value between Ø and 15 and click **Update Unit**.

Line In Continuous Stream forces the AM36Øa to stream audio to the network continuously, regardless of the state of the I/O 1 input on the rear. Check this box to force continuous streaming; clear this checkbox to require the I/O 1 on the rear connector to control streaming to the network.

VOX Enable provides positive control of the voice activation feature. When VOX is enabled, any sound from the microphone or line inputs that exceeds the **VOX Threshold** level will trigger the sending of packets to the IP network. Check this box to enable VOX using the values entered in the following fields; clear this checkbox to disable VOX and require the hardware logic inputs to trigger audio streaming.

VOX Threshold sets the audio level above which the AM36Øa will begin streaming audio to the network without requiring a hardware digital signal (COS or I/O 1) to go active. Type a number between Ø and 5Ø that provides the desired operation. A value of Ø is the most sensitive setting while 5Ø is the least sensitive. NOTE that VOX operation will prevent proper operation of channel patches on soft console systems.

VOX Timeout enables the VOX feature and sets the hang time during which the AM36Øa will continue to stream RTP packets to the network after the audio level falls below the **VOX Threshold** level. Set this to a value between Ø.1 and 25.5 seconds that eliminates choppy operation with the type of audio that you are streaming. When this value is set to Ø (zero) the VOX feature will be disabled for backward compatibility.

Speaker Volume displays the current setting of the Speaker output level. To set this Volume level electronically, type in a value between Ø and 15 and click **Update Unit**.

Line Out Level displays the current setting of the Line Output level. To set this level electronically, type in a value between Ø and 15 and click **Update Unit**.





Save Your Changes

Update Unit button — When you are satisfied with all the settings that appear on the **Global Configuration** page, be sure to click on the button labeled **Update Unit** at the top of the page. Your new settings take effect immediately.

Undo Edits button — To revert back to the previously saved values while editing click the button labeled **Undo Edits** at the top of the page.

Reboot checkbox — To reboot, check the **Reboot** box and click the **Update Unit** button. Any changes are saved before the reboot is executed.

Volume Knob

To adjust audio levels using the Volume knob on the front of the unit, push in the knob to select the desired audio line. Each push of the knob advances the selection to the next audio line in sequence, as indicated by the lamps on the front. Once the desired line is selected, turn the knob slowly clockwise to increase the audio level or counter-clockwise to reduce the audio level. After a short period of time the selector will revert back to Speaker. These manual adjustments will not be saved; upon reboot the stored values typed into the Module Configuration screen will be loaded.



STATUS

Status

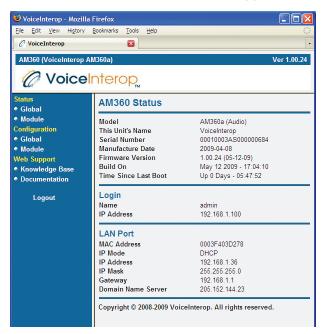
The AudioMate 36Ø family of IP gateways provide status reports to make adjustments and diagnostics easier.

The AudioMate 360a provides the means to view the status of the digital input and output signals as well as to view the current settings of the transmit and receive audio levels.

Global Status

The Global Status page displays various parameters of the common interface board.

In the menu pane on the left, locate the **Status** heading and click the link **Global** under it. The screen labeled **AM36Ø Status** should appear in the right pane.



Some noteworthy fields:

Model shows the particular type of AudioMate 360 unit that you are managing.

This Unit's Name displays text that was stored to identify this unit.

Serial Number uniquely identifies this unit.

Login

IP Address shows the address of YOUR computer from which you are managing this unit.



LAN Port

MAC Address shows the Ethernet address of the unit's LAN interface. This uniquely identifies the unit on your network and can be useful in differentiating among several AudioMate 360 units on the same network.

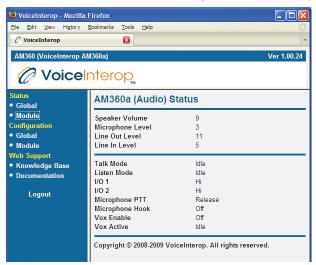
IP Mode shows whether the unit received its address from a DHCP server or is configured with a static IP address.

IP Address shows the currently assigned IP address regardless of how it was obtained.

Module Status

The Module Status page displays the audio level settings and the state of the various digital inputs and outputs. Refresh the page to see the current state.

In the menu pane on the left, locate the **Status** heading and click the link **Module** under it. The screen labeled **AM36Øa** (**Analog**) **Status** should appear in the right pane.



Speaker Volume, Microphone Level, Line Out Level, and **Line In Level** display the current setting of their respective audio levels. These can be changed in the **Module Configuration** screen.

Talk Mode when **Active** indicates that voice packets were being streamed to the LAN port at the moment when this page was refreshed. This can be triggered on demand by the Microphone PTT switch, I/O 1 input, VOX activation, or full time by checking either the **Microphone Continuous Stream** or **Line In Continuous Stream** checkboxes.

Listen Mode when **Active** indicates that the LAN port was receiving voice packets addressed to this unit at the moment when this page was refreshed.

I/O 1 indicates the state of this digital input line on the rear connector. When Hi it is idle and no RTP voice packets are streamed to the network. When it shows Low it is active and RTP packets will be streamed from the Line Input to the network.





I/O 2 indicates the state of this digital output line on the rear connector. When Hi it is idle. When it shows Low it is active, indicating that audio is available on the Line Output jack. This might be used to drive a mic precedence (mute) input on a PA amplifier or to trigger a circuit that activates indicator lamps to draw attention to an announcement.

Microphone PTT shows the state of the PTT switch on the Microphone input.

Microphone Hook is not used.

VOX Enable shows the configured state of the voice detector. When **Off** the hardware inputs must be used to initiate streaming of packets to the IP network. When **On** the voice detector will use the other VOX settings to trigger the streaming of packets.

VOX Active displays **Active** if the voice detector was triggered at the moment that this status page was refreshed. Otherwise it shows **Idle**.



SPECS

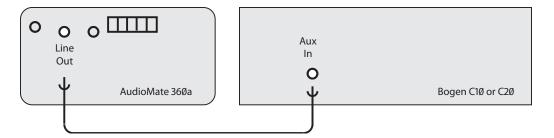
AudioMate 360a Specifications

Physical		
Dimensions	1.75 in. (4.5 cm) H x 3.6 in. (9.2 cm) W x 6.0 in. (15.3 cm) D	
Weight	Ø.8 pounds (Ø.4 kg)	
Electrical		
Input Voltage	12 Volts d.c. nominal, 16 Volts d.c. max., 10 Volts d.c. min.	
Input current	35Ø mAmps nominal, 1 Amp peak	
Audio Interface		
Line input	2.2 Volts peak-peak nominal, 5 Volts peak-peak max. @ 10K Ohms	
Line output	2.2 Volts peak-peak nominal, 5 Volts peak-peak max. @ 100 Ohms	
Microphone input	200 mVolts peak-peak @ 1K Ohms	
Speaker output	1 Watt @ 8 Ohms w/3% t.h.d.	
Digital I/O		
Output	12 Volts @ 500 mAmp max., open drain; 1 active low output indicating audio streaming from IP network	
Input	high > 3 Volts (18 Volts max.); 1 active low input to begin streaming to IP network	
L.A.N.	10/100-BaseT Ethernet RJ-45 port	
Serial	USB-mini B diagnostic port	
Digital audio		
CoDecs	G.711 μ-law (64kbps), G.711 A-law (64kbps), GSM 6.10 (13kbps)	
Environmental		
Operating temperature	Ø to +5Ø Centigrade	
Storage temperature	-2Ø to +6Ø Centigrade	
Miscellaneous		
Indicators	Activity: OK, Link, Tx, Rx; Levels: Speaker, Mic, Line in, Line out	
Features	Firmware upgrade over Ethernet; web browser configuration screens (no jumpers); telnet diagnostic output, USB diagnostic and basic configuration utility	
Ports	Microphone RJ-45 (on front panel); line in, line out, speaker all 3.5mm phone jacks; 5-pin inline digital I/O; LAN RJ-45; 12 Volt coaxial power jack; USB-mini B for diagnostics	



A

Appendix A - Interface ExamplesPublic Address (Paging) Amplifier



Connection to a typical PA amplifier is simple and easy. Among the leading manufacturers, such as Bogen and Cooper-Wheelock, specifics of the connections vary; in the example depicted a Sound Blaster cable from 3.5mm phone plug to RCA plug will suffice. In most cases the Line Out of the AM36Øa connects to a line-level input of the PA amp, usually labeled Aux or Line.

LMR Remote Control

The AM36Øa can be employed as a simple radio remote control console talking to an AM36Ør on a point-to-point IP unicast link or as part of a larger WAVE installation. For use with WAVE, configure the AM36Øa as normal with a multicast group RTP address and port, create a matching Multicast Trunk in WAVE, then bridge this new WAVE channel with the desired standard channel via an Advanced Session.

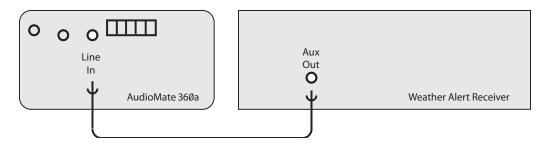
When the AM360a is outfitted with a Kenwood KMC-27B hand PTT microphone or KMC-9C desk PTT microphone and an 8-Ohm speaker it becomes another link into your VoIP communications network.

Microphone Jack Pin-Outs			
Pin	Name	Remarks	
1	BLC	n/a	
2	Switched Power	Phantom power, 12 Volts	
3	GND	Ground return	
4	PTT	Push-to-talk	
5	Mic.G	Mic audio low / ground	
6	Mic.S	Mic audio hi	
7	Hook	Hook switch input (not implemented)	
8	СМ	n/a	
Pin 1 is on the right when looking at the front of the AM360a			



Weather Radio Channel

Public safety and business entities that have fleets that work outdoors can benefit from keeping decision makers abreast of changing weather conditions. A single inexpensive weather receiver or radio scanner can be connected to the Line In of the AM36Øa to provide vital weather reports and severe weather alerts to everyone on your VoIP communications network.



Because this is a one-way audio application, either the continuous streaming or the VOX feature of the AM360a can be put to good use.

Digital I/O Connections

Digital I/O on Rear Apron		
Pin	Function	
I/O 1	(COS) input, pull low to begin streaming line in to network	
I/O 2	(PTT) output, goes low when RTP stream received from network	
RS485+	RS485 serial data, not implemented	
RS485-	RS485 serial data, not implemented	
GND	Ground	





Appendix B - Multicast Addressing NotesLimitations on Multicast Group Address Choices

Note 1: Avoid addresses of the form 224.0.0.x

Traffic to addresses of the form 224.0.0.x is often flooded to all switch ports. This address range is reserved for link-local uses. Many routing protocols assume that all traffic within this range will be received by all routers on the network. Hence (at least all Cisco) switches flood traffic within this range. The flooding behavior overrides the normal selective forwarding behavior of a multicast-aware switch (e.g. IGMP snooping, CGMP, etc.).

Note 2: Watch for multicast address ambiguity

32 IP multicast addresses are mapped onto each Ethernet multicast address. A receiver that joins a single IP multicast group implicitly joins 31 others due to this overlap. The overlap occurs in the 5 high-order bits, so it's best to use the 23 low-order bits to make distinct multicast streams unique. For example, IP multicast addresses in the range 239.0.0.0 to 239.127.255.255 all map to unique Ethernet multicast addresses. However, IP multicast address 239.128.0.0 maps to 239.0.0.1, etc.

Note 3: Avoid x.0.0.y and x.128.0.y

Combining notes 1 and 2, it's best to avoid using IP multicast addresses of the form $\mathbf{x.0.0.y}$ and $\mathbf{x.128.0.y}$ since they all map onto the range of Ethernet multicast addresses that are flooded to all switch ports.

Note 4: Watch for address assignment conflicts

IANA administers multicast addresses on the world-wide Internet. Potential conflicts with Internet multicast address assignments can be avoided by using (private) administratively scoped addresses. Such addresses can be safely used on a network connected to the Internet without fear of conflict with multicast sources originating on the Internet. Administratively Scoped addresses are roughly analogous to the unicast address space for private internets. Site-local multicast addresses are of the form 239.255.x.y, but can grow down to 239.252.x.y if needed. Organization-local multicast addresses are of the form 239.192-251.x.y, but can grow down to 239.x.y.z if needed.

The address range 239.252.1.1 through 239.255.254 is preferred to avoid multicast packet propagation outside of the private network.

AudioMate 360

More info at: cisco.com 29west.com smallnetbuilder.com

- » See also Cisco's Guidelines for Enterprise IP Multicast Address Allocation paper available at www.cisco.com
- » See also "Multicast Address Assignment" at www.29west.com
- » See also Doug Reid's 3 part series "Multicasting and the Small Network" at www.smallnetbuilder.com





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Appendix C - WAVE Configuration

WAVE (Wide Area Voice Environment) from Twisted Pair Solutions provides the glue to tie together your various IP voice communications endpoints. The AudioMate 36Ø family of IP voice gateways works seamlessly with WAVE to extend your communications options and enhance interoperability among disparate devices.

Multicast groups are preferred for WAVE installations to allow many soft consoles to participate in the same communication channels without increasing the bandwidth consumed by each console. Deploying multicast links to each AudioMate 36Ø device allows for seamless failover if that option is implemented in WAVE. Any participating media server listens to the multicast streams and picks up where the failed media server dropped off.

However, in some networks the only option is unicast transmission. This forces the AudioMate 36Ø units to be configured as point-to-point trunk links and prevents their streams from being picked up automatically by a failover server. In other words, if the target media server fails at the other end of the point-to-point pipe the packets have nowhere to go. As long as these limitations are acceptable, unicast point-to-point links can be employed.

Multicast Configuration AM360

Open a web browser and point it to the management IP address of the AudioMate 360 device to be configured. Log in with user admin and password password or the credentials that you have configured.

Under Configuration click Global.

In the field RTP Remote IP type the desired multicast group address, following the guidelines in Appendix B. Addresses in the 239.255.x.y block are recommended for private local area networks. For RTP Remote Port and RTP Local Port type a port number between 16384 and 65534; both fields should be identical numbers. For Codec select G.711u; if this network segment will not support about 100kbps streams then select GSM. Click the button Update Unit to save your changes.

WAVE

CHANNELS (WAVE 3)

Log in to WAVE. Click **Channels** -> **Channels**, then click the **Add Channel** button. From the dropdown listbox select **Multicast Trunk**, then click the **Next** button. In the **Name** field type a meaningful name. Tab to the **Description** field and type a meaningful description. Select a group from the **Group** listbox.

In the IP address field type the exact same multicast group address that you configured into the AM36Ø above. Tab over to the Port field and type the exact same port number that you configured into the AM36Ø above. Check the box labeled Use the same IP address and port as 'Audio Receive'. In the Codec listbox select the same codec as was selected in the AM36Ø above; if you choose GSM then set the Scale to 1. All remaining settings may be left at their defaults.

Click the **Save** button to save your changes. If there are no other configuration changes to be entered, click the button **Commit Changes**.





CHANNELS (WAVE 4)

Log in to WAVE. Click **Media Servers** -> **Interfaces**, then click the **Add Interface** button. From the dropdown listbox select **VoiceInterop AM360** and click the **Next** button. In the **Name** field type a meaningful name. Tab to the **Description** field and type a meaningful description.

In the **Gateway IP Address** field type the management address of the AM360 that will be associated with this endpoint. In the **RTP Remote IP address** field type the exact same multicast group address that you configured into the AM360 above. Tab over to the **RTP Remote Port** field and type the exact same port number that you configured into the AM360 above. In the **RTP Local IP address** field type the exact same multicast group address that you configured into the AM360 above. Tab over to the **RTP Local Port** field and type the exact same port number that you configured into the AM360 above.

In the **Codec** listbox select the same codec as was selected in the AM36Ø above; if you choose GSM then set the **Scale** to 1.

Click the **Save** button to save your changes. If there are no other configuration changes to be entered, click the button **Commit Changes**.

Sessions (WAVE 3)

If you already have the desired standard user channel configured, click **Sessions** -> **Sessions**, then click the **Add Session** button.

In the dropdown listbox select **Advanced**, then click the **Next** button.

Type a meaningful name in the **Name** field. Tab to the **Description** field and type a meaningful name. Select the desired group from the **Group** listbox and click the **Save** button to reveal the full Session configuration screen.

Leave the Conference ID / DN field blank and clear the checkbox labeled Treat as the Dial-In DN. Click the Add Servers button and select the desired media servers that will handle this session. Click the Add Participant button and add the multicast trunk channel from above and the desired standard user channel that will be bridged with this AM360 device.

Click the button **Save and Commit changes**. Your AudioMate 360 device is now bridged via multicast with the desired WAVE Standard channel. Test the configuration using an AudioMate Communicator soft console.

WAVE 4

The AudioMate 360 is integrated into WAVE 4 as a media server interface. Add an AM360 to a session as an interface.

Sessions (WAVE 4)

To configure a session in WAVE version 4, proceed as directed for WAVE 3 installations. The only difference comes when adding participants to the session. After adding the standard user channel as indicated for WAVE 3, click the **Add Participant** button again. In the dropdown listbox select **VoiceInterop AM360**, then click the **Next** button. Select the desired AM360 by checking the box next to it, then click the **Add Selected Participants** button.

Unicast Configuration AM360

Open a web browser and point it to the management IP address of the AudioMate 360 device to be configured. Log in with user admin and password password or the credentials that you have configured.





Under Configuration click Global.

In the field RTP Remote IP type the unicast address of the target media server. For RTP Remote Port and RTP Local Port type a port number between 16384 and 65534; both fields should be identical numbers. For Codec select G.711u; if your network will not support about 100kbps streams then select GSM. Click the button Update Unit to save your changes.

WAVE

CHANNELS (WAVE 3)

Log in to WAVE. Click **Channels** -> **Channels**, then click the **Add Channel** button. From the dropdown listbox select **Point-to-point Trunk**, then click the **Next** button. In the **Name** field type a meaningful name. Tab to the **Description** field and type a meaningful description. Select a group from the **Group** listbox.

In the **Trunk Side A IP address** field type the IP address of the target media server. In the **Port** field type the same port number that you configured into the AM36Ø above.

In the **Trunk Side B IP address** field type the IP address of the AM36Ø above. Tab over to the **Port** field and type the exact same port number that you configured into the AM36Ø above. In the **Codec** listbox select the same codec as was selected in the AM36Ø above; if you choose **GSM** then set the Scale to 1. All remaining settings may be left at their defaults.

Click the **Save** button to save your changes. If there are no other configuration changes to be entered, click the button **Commit Changes**; else, configure Sessions as above.

CHANNELS (WAVE 4)

Log in to WAVE. Click Media Servers -> Interfaces, then click the Add Interface button. From the dropdown listbox select VoiceInterop AM36Ø and click the Next button. In the Name field type a meaningful name. Tab to the Description field and type a meaningful description.

Remote vs Local

The terms Remote and Local are relative to the device that you are configuring. In WAVE, the media server is the Local address while Remote refers to the AudioMate 360 unit.

In the **Gateway IP Address** field type the management address of the AM36Ø that will be associated with this endpoint. In the **RTP Remote IP address** field type the management IP address of the AM36Ø above. Tab over to the **RTP Remote Port** field and type the exact same port number that you configured into the AM36Ø's field labeled 'RTP Local Port'. Leave the **RTP Local IP address** field blank to use the binding IP address of the target media server. Tab over to the **RTP Local Port** field and type the exact same port number that you configured into the AM36Ø's field labeled 'RTP Remote Port'.

In the **Codec** listbox select the same codec as was selected in the AM360 above; if you choose **GSM** then set the **Scale** to 1.

Click the **Save** button to save your changes. If there are no other configuration changes to be entered, click the button **Commit Changes**.

The session that bridges these unicast channels are configured in a procedure similar to bridging multicast channels in WAVE version 4.