

Tonmind IP Speaker User Manual



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1. Overview

Tonmind IP speaker have different shape and design to fulfill the installation in indoor and outdoor environment, they are well compatible with SIP & ONVIF protocol that are able to be used in VoIP and security field. Up to 10 RTP multicast address enable to arrange different paging solutions, alarm in and HTTP URL are able to combine with alarm system. Pre-recorded message and schedule broadcasting to meet various paging demands. The 48K OPUS Audio Codec enables excellent sound quality, to make announcement, play background music, security alarm in school, factory and hospital, etc.



2. Web Configuration

Web configuration includes complete function setting, after the device and PC are connected to a same network, type the device's IP address in a web browser, the default IP address is 192.168.5.200, then log in with defaulted username and password as below, and there are different language options.

Username: admin

Password: tm1234

Login
IP Speaker

Username

Password

Language

Sign in
Cancel

[Forgot Password?](#)

2.1 Status

Show out device time, firmware version, free space and SIP accounts status, and also MAC, IP address and gateway etc.

Status

Device Time	2024-10-17 02:38:35
Device ID	50359289708D641C
Firmware Ver	CS20-V3.3.36
Free Space	3576KB
SIP1 Status	NONE
SIP2 Status	NONE

Network

MAC Address	A2:C0:A4:20:29:4C
IP Address	192.168.2.101
Subnet Mask	255.255.255.0
Gateway	192.168.2.1
Primary DNS	192.168.5.1
Secondary DNS	192.168.2.1

Refresh

2.2 Basic

2.2.1 Date/ Time

Two update modes for time: NTP and local time.

- NTP: set time zone, NTP sever and interval, then save the configuration.

Date/Time

Device Time	2024-10-17 02:42:47		
Update Mode	NTP	▼	
TimeZone	GMT+08:00	▼	
NTP Server	pool.ntp.org		
NTP Interval	10	Minutes	

Save

- Local time, follow the PC time.

Date/Time

Device Time	2024-10-17 02:42:47		
Update Mode	LocalTime	▼	
LocalTime	2024-10-17 11:31:07		

Save

2.2.2 Network

- DHCP, IP address will be created automatically by DHCP server
- Status IP address, it could set up as required of IP Address, Subnet Mask, Gateway, Primary DNS, and Secondary DNS.

Network

DHCP
 Static IP Address

IP Address:
 Subnet Mask:
 Gateway:
 Primary DNS:
 Secondary DNS:

[Save](#)

2.2.3 Network Advanced

Set Http or Https or both, and the ports of Http, Https and RTSP. And VLAN configure, able for VLAN ID, VLAN IP, VLAN Netmask, and VLAN Gateway setting.

Network Advanced (*Take effect after device is restarted!)

Http/Https:

Http Port: (80, 1025~65534)
 Https Port: (443, 1025~65534)
 RTSP Port: (554, 1025~65534)

VLAN Enable:

VLAN ID (1~4094):

VLAN IP:
 VLAN Netmask:
 VLAN Gateway:

[Save](#)

2.3 ONVIF

- Select Enable ONVIF, then the device is able to be searched by ONVIF VMS, the password and user name is same like IP speaker log in user name and password.
- WAN NAT, do ports forward on router to ensure that data can be transmitted correctly between the server and the public Internet, then come to speaker web and turn on the option **WAN NAT**, insert the IP address, HTTP Port and RTSP Port.

ONVIF

ONVIF Enable

WAN NAT

WAN IP Address

WAN HTTP Port

WAN RTSP Port

Save

2.4 SIP Account

2.4.1 SIP Set

Each speaker has two SIP accounts, put SIP server extension messages into the blanks and save the configuration, then you can check if it registers successfully or not.

User Name	User account, provided by SIP server
Auth ID	SIP service subscriber's ID used for authentication.
Password	Account password provided by SIP server
Display Name	SIP service subscriber's name
Server Host	SIP server address
Server Port	SIP port, default to be 5060
Outbound Proxy	It is used to process signals and help data streams to go through firewall or NAT if there have.
Expire Time	Set the expire time of registered account information
Ringtone	5 system ringtones and 10 users upload media files
Auto Answer	answer immediately and answer delay when a calling income
Incoming Notify	Put an input URL, when a incoming call ringing, URL take effect, include to play HTTP audio stream

Answer Notify	Put an input URL, when a incoming call answered, URL take effect, include to play HTTP audio stream
---------------	---

SIP Set

Account	<input type="text" value="Account 1"/>	NONE
User Name	<input type="text"/>	
Auth ID	<input type="text"/>	
Password	<input type="text"/>	
Display Name	<input type="text"/>	
Server Host	<input type="text"/>	
Server Port	<input type="text"/>	
Outbound Proxy	<input type="text" value="Disable"/>	
Expire Time	<input type="text" value="180"/>	Seconds
Ringing Tone	<input type="text" value="bell1"/>	
Auto Answer	<input type="text" value="Answer Immediatly"/>	
Incoming Notify	<input checked="" type="checkbox"/>	
Http URL	<input type="text"/>	
Answer Notify	<input checked="" type="checkbox"/>	
Http URL	<input type="text"/>	

2.4.2 SIP Advanced

Now IP speakers support to work under transport protocols of UDP, TCP and TLS, with the encryption of SRTP option.

Then if want to support SIP P2P (Peer-to-peer SIP), then click to make it work. If it takes place on a local network, all that's needed are the SIP addresses of the user agents. A typical SIP address in this case would be SIP:<local IP>, like sip:192.168.5.200.

SIP Advanced

SIP Protocol	UDP
Encryption	SRTP
SIP P2P Enable	<input checked="" type="checkbox"/>

Save

2.5 Audio

2.5.1 Codec

Under Audio, select at least one audio codec with the desired audio quality, there are four options.

Codec

Codec Setting	<input checked="" type="checkbox"/> OPUS
	<input checked="" type="checkbox"/> G.722
	<input checked="" type="checkbox"/> G.711U
	<input checked="" type="checkbox"/> G.711A

2.5.2 Speaker

- Volume: Speaker out volume
- Amp Auto OFF: It's set defaulted YES, to turn of the build in amplifier if do not play audio.
- Jitter buffer: Make the audio playing more stable.
- HPF: high pass filter, it is to filter the frequency under 150HZ.
- NR: noise reduction, it is able to reduce the noise from audio input to improve sound quality when enable, it is calculated and simulated from the Chip.

Speaker

Volume (0-100)	<input type="text" value="60"/>	
Amp Auto OFF	<input type="text" value="YES"/>	▼
Jitter Buffer (60 - 2000)	<input type="text" value="360"/>	ms
HPF	<input type="checkbox"/>	
NR	<input type="checkbox"/>	

2.5.3 MIC



- Gain: There are four different gain level from none to high.
- Volume: Build in MIC out volume
- AEC: Acoustic Echo Cancellation
- AGC: Automatic Gain Control
- HPF: high pass filter, it is to filter the frequency under 150HZ.
- NR: noise reduction, there are 3 level to reduce Premeter noise from speaker Mic side for listen. It is calculated and simulated from the Chip.

MIC

Gain	<input type="text" value="None"/>	▼
Volume (0-100)	<input type="text" value="100"/>	
AEC	<input checked="" type="checkbox"/>	
AGC	<input checked="" type="checkbox"/>	
AGC Gain Level	<input type="text" value="High"/>	▼
HPF	<input type="checkbox"/>	
NR	<input checked="" type="checkbox"/>	
NR Level	<input type="text" value="1"/>	▼











Save

2.6 Media File

Click  to listen in PC, and click  to listen in speaker side.











- System File

Five system audio clips

System File		
#	Name	
1	bell1	 
2	bell2	 
3	bell3	 
4	bell4	 
5	bell5	 

- User file

10 audio clips upload bases on customers' demands, the free space is about 3800kb.

User File (3796KB free)			
#	Name	File	
1	userfile1	<input type="button" value="选择文件"/> 未送	
2	userfile2	<input type="button" value="选择文件"/> 未送	
3	userfile3	<input type="button" value="选择文件"/> 未送	
4	userfile4	<input type="button" value="选择文件"/> 未送	
5	userfile5	<input type="button" value="选择文件"/> 未送	
6	userfile6	<input type="button" value="选择文件"/> 未送	
7	userfile7	<input type="button" value="选择文件"/> 未送	
8	userfile8	<input type="button" value="选择文件"/> 未送	
9	userfile9	<input type="button" value="选择文件"/> 未送	
10	userfile10	<input type="button" value="选择文件"/> 未送	

2.7 Alarm

Two kind of alarm trigger, one is connected to speaker alarm in IO connector and the other one is HTTP API.

2.7.1 Alarm In (IO connection)

- File Enable: play pre-recorded audio
- Sip Enable: enable SIP extension and SIP P2P call
- Http Stream Enable: play HTTP audio stream, like http://listen.livestreamingservice.com/181-greatoldies_128k.mp3

Alarm In

File Enable	<input checked="" type="checkbox"/>	
Play File		<input type="text" value="bell1"/> ▶
Cycle Mode		<input type="text" value="Once only"/>
Sip Enable	<input checked="" type="checkbox"/>	
Sip Account		<input type="text" value="Account 1"/>
Sip Number		<input type="text" value="6688"/>
Http Stream Enable	<input checked="" type="checkbox"/>	
Http Stream URL		<input type="text" value="http://listen.livestreamingservice.com/181-greatoldies_128k"/>

2.7.2 HTTP URL

- (1) Enable Play URL Enable
- (2) Speaker will be able to receive supported HTTP URL

Http URL

Play File Enable Example1: `http://192.168.2.100/api/play?action=start&file=bell1`Example2: `http://192.168.2.100/api/play?
action=start&file=userfile1&mode=once&volume=10`Example3: `http://192.168.2.100/api/play?
action=start&file=userfile1&mode=multiple&count=10&volume=20`Example4: `http://192.168.2.100/api/play?
action=start&file=userfile1&mode=duration&count=10&volume=30`Example5: `http://192.168.2.100/api/play?action=stop`Example6: `http://192.168.2.100/api/play?
action=startstream&stream=http://xxxxxx`Example7: `http://192.168.2.100/api/play?action=stopstream`

2.8 Schedule

It is widely use in school, factory and office, make a regular bell, announcement and alarm, support 10 schedule set up.

- Enable the Schedule Enable
- Schedule Name
- Loop Type: Once, Daily and weekly
- Action Time
- Action Type: Start or Stop
- Play File: Pre-recorded audio clips
- Cycle Mode
- Http Stream URL: play HTTP audio stream, like http://listen.livestreamingservice.com/181-greatoldies_128k.mp3

Schedule Add/Edit

Schedule Enable	<input checked="" type="checkbox"/>
Schedule Name	<input type="text" value="Tonmind"/>
Loop Type	<input type="text" value="Daily"/> ▼
Action Time	08:00 🕒
Action Type	<input type="text" value="Start"/> ▼
Play File	<input type="text" value="bell1"/> ▼ 🎵
Cycle Mode	<input type="text" value="Once only"/> ▼
Http Stream URL	<input type="text" value="http://listen.livestreamingservice.com/181-greatoldies_128k"/>

2.9 RTP Multicast

Support 10 RTP addresses, please note that: port numbers do not use continuous numbers when setting the same RTP addresses. Use discontinuous numbers. eg:

239.255.1.2:8000, 239.255.0.1:8001, 239.255.0.1:8002 (×)

239.255.0.1:8000, 239.255.0.1:8002, 239.255.0.1:8004 (√)

- Multicast address range: 224.0.0.0-239.255.255.
- Ports range: 1024-65536
- Use IP Tool, Audio Manager, PA Lite and PA Pro to do RTP multicast.

RTP Multicast

Priority	IP Address (e.g. 239.255.0.1:5004)
1	<input type="text" value="239.255.0.0:8000"/>
2	<input type="text"/>
3	<input type="text"/>
4	<input type="text"/>
5	<input type="text"/>
6	<input type="text"/>
7	<input type="text"/>
8	<input type="text"/>
9	<input type="text"/>
10	<input type="text"/>

2.10 Firewall

2.10.1 Firewall Rules

- Name
- Rule Type: IP address or MAC
- Protocol: ALL or TCP, UDP
- IP Address or Mac: like 192.168.5.200
- Net Mask:
- Action: Accept or not.

Firewall Add/Edit

Enable	<input checked="" type="checkbox"/>
Name	<input style="width: 90%;" type="text"/>
Rule Type	<input type="text" value="IP"/> ▼
Protocol	<input type="text" value="ALL"/> ▼
IP Address	<input style="width: 90%;" type="text"/>
Net Mask	<input style="width: 90%;" type="text"/>
Action	<input type="text" value="ACCEPT"/> ▼

2.10.2 Automatic Defense

Set to protect from the TCP and UDP port or ICMP.

Automatic Defense Add/Edit

Enable	<input checked="" type="checkbox"/>
Name	<input style="width: 90%;" type="text"/>
Protocol	<input type="text" value="TCP"/> ▼
Port Range	<input style="width: 45%;" type="text"/> - <input style="width: 45%;" type="text"/>
Rate (1-10000)	<input style="width: 45%;" type="text"/> /s

2.11 Auto Provision

Three ways to set up IP speaker auto provision, it is MAC-based configuration provisioning, support third-party server storing configuration files of DHCP, PnP, TFTP, FTP and HTTP.

2.11.1 DHCP option

set up a DHCP server first and then choose the option code, there three choices: option66, option43, Custom option, then follow the way to set the routine of configure file and run DHCP server.

DHCP Option

DHCP Option Setting Option 66 ▾

2.11.2 PnP

Set up a PNP server, then follow the way to set the routine of configure file and run PNP server.

PnP

Enable PnP

PnP Server

PnP Port

PnP Transport UDP ▾

PnP Interval (1~99)Hour

2.11.3 Static Provisioning Server

Supports three kinds of protocol: TFTP, FTP and HTTP. Prepare a TFTP, FTP and HTTP server, and set up server address and the saving path of auto provision file, if the server needs authentication information, set up and remember the username and password.

Static Provisioning Server

Update Mode Update After Reboot ▾

Update Interval (1~99)Hour

Server Address

Protocol Type TFTP ▾

Username

Password

Save

2.12 System

2.12.1 Maintain

- Log: speaker running situation record
- Reboot
- Reset
- Upgrade

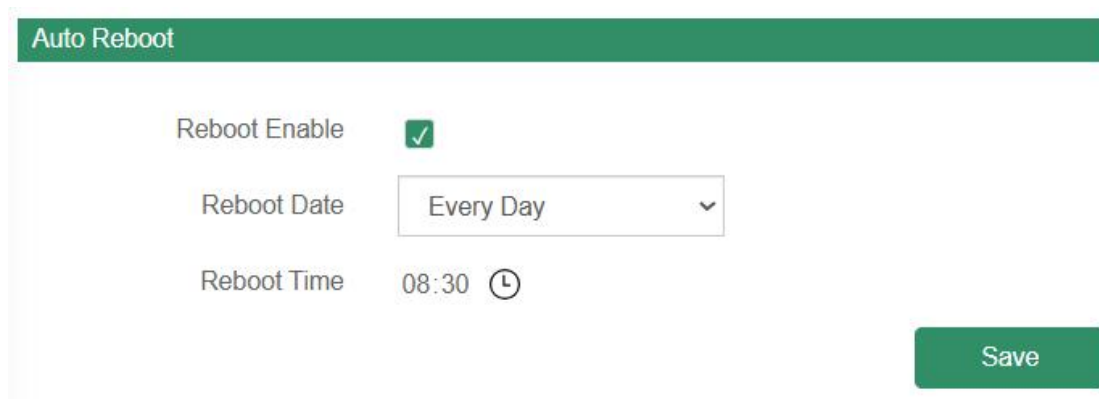
How to upgrade IP Speaker firmware version in web interface?

- (1) Select the latest version firmware: xxx-bin.
- (2) Click upgrade, it would require about 20s to finish process.
- (3) Back to speaker login web interface.



2.12.2 Auto Reboot

Set to reboot as desired time.



2.12.3 Security

Set a new user name and password, save the configuration and restart login.

Security

User Name	<input style="width: 90%;" type="text" value="admin"/>
Password	<input style="width: 90%;" type="password" value="....."/>
New User Name	<input style="width: 90%;" type="text" value="Tonmind"/>
New Password	<input style="width: 90%;" type="password" value="....."/>
Confirm Password	<input style="width: 90%;" type="password" value="....."/>

3. IPTool Configuration

Apart from Web configuration, IPTool is the other option that configure quickly basic information such as SIP account setting, volume setting, RTP Multicast setting, upgrade. Please follow below steps.

- (1) Download IPTool in <https://www.tonmind.com/category/downloads/5>
- (2) Run IPTool, select correct networking from "Network", and then "Search" the devices and do setting.

