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Setting up a Deskphone



Written by Kamila
Updated over a week ago

How do I get my connection details?

1. Before you set up your deskphone, you first must register a new personal contact and get your connection details.

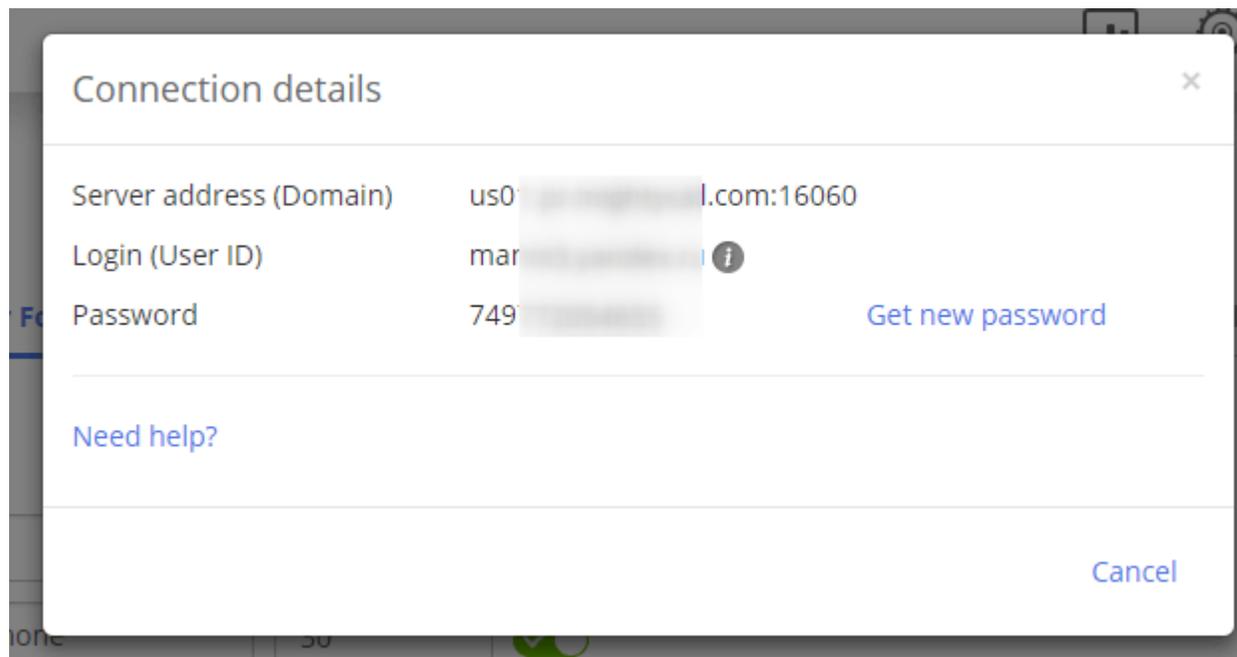
Go to your account => Profile ==> My Forwarding Numbers to start the process.

The screenshot shows a web browser window with the URL <https://panel.mightycall.com/MightyCall/Profile#/voice/contact>. The page title is "My Forwarding Numbers" under the "Profile" section. The interface includes a sidebar with navigation icons and a main content area with a table of forwarding numbers. A dropdown menu is open in the top right corner, showing options like "Profile", "Account", "Billing information", "Payment History", "Subscription", and "Log out". The "Profile" option is highlighted with a red box. In the table, the "SIP Sip Phone" row has a "Get connection details" button highlighted with a red box.

Phone number (contact)	Timeout	Enabled
APP Mobile App (Incoming calls)	30	<input checked="" type="checkbox"/>
US +1 (929) 363-5577	30	<input checked="" type="checkbox"/>
WEB MightyCall Webphone	30	<input checked="" type="checkbox"/>
SIP Sip Phone	30	<input checked="" type="checkbox"/>

[+ Add Forwarding number](#)

Ring the numbers Sequentially Simultaneously



1. Click on the Add softphone (SIP account) link.
2. Make sure to save your changes!
3. Then click on the “Get Connection details” link.
4. Here you can view your connection details. You can use these credentials to create your SIP account for a third-party SIP phone (see a list below).

- **Server address (domain)**
- **Login (user ID)**
- **Password**

Cisco SPA303

How to Use a SIP Phone:

1. Connect the AC adapter cord to the SW connector on your phone.
2. Wait until the menu screen appears. Then, press the Menu button.



Static IP connection:

1. Open the Network menu, choose “WAN Connection Type”, and change its value to “Static IP”.
2. Fill the “Non-DHCP IP Address”, “Non-DHCP Subnet Mask”, “Non-DHCP Default Route”, “Non-DHCP DNS 1” and “Non-DHCP DNS 2” fields with credentials that were provided by

- your internet provider.
3. Return to the main menu and then reboot your phone.
4. After the reboot is complete, open the Network menu and check the field labeled “Current IP”. Here you should find the IP address from step 1.
5. Enter this IP as the URL in the browser on your computer. Log in as the administrator by clicking on the “Admin login” button.

The screenshot displays the Cisco SPA303 Configuration Utility interface. At the top right, there is a navigation bar with 'Admin Login' highlighted in a red box, along with 'Basic' and 'Advanced' options. The main content area is organized into several sections:

- System Information:** Connection Type: Static IP; Current IP: 80.254.16.158; Current Gateway: 80.254.16.157.
- Reboot History:** Lists several reboot events with reasons like 'User Triggered'.
- Product Information:** Product Name: SPA303; Software Version: 7.4.9c; Serial Number: CCO161508GD.
- Phone Status:** Current Time: 10/4/2017 09:55:32; Elapsed Time: 00:56:54.
- Ext 1 Status:** Registration State: Not Registered.
- Ext 2 Status:** Registration State: Not Registered.

At the bottom of the interface, there are buttons for 'Undo All Changes' and 'Submit All Changes'.

6. Choose the “Ext 1” tab, then set the value of the “Line Enable” field to YES. Fill in the “SIP Port”, “Proxy”, “User ID”, “Password” fields.

You can find your credentials here: Your Profile > My Forwarding numbers > Get connection details.

Small Business
CISCO SPA303 Configuration Utility User Login basic | advanced

Voice Call History Personal Directory

Info System SIP Regional Phone User

Ext 1 Ext 2 Ext 3

General
 1 Line Enable: Restrict MWI:

NAT Settings
 NAT Mapping Enable: NAT Keep Alive Enable:

SIP Settings
 2 SIP Port: SIP Debug Option:

Call Feature Settings
 Message Waiting: Default Ring:
 Mailbox ID: User ID with Domain:
 Auto Ans Page On Active Call: Feature Key Sync:

Proxy and Registration
 3 Proxy:
 Register: Make Call Without Reg:
 Register Expires: Ans Call Without Reg:

Subscriber Information
 Display Name: 4 User ID:
 5 Password: Use Auth ID:
 Auth ID:

Undo All Changes Submit All Changes

7. Select “Submit all Changes” and wait while your phone is rebooting.

8. Choose the “Info” tab on the administrator panel and check the field labeled “Registration State” in the “Ext 1 Status” section. There, you should find the status set to “Registered”.

NAT connection:

1. Open the “Network” menu, choose “WAN Connection Type”, and change its value to “DHCP”. Check the field labeled “Current IP”. Here you should see your phone’s IP address.
2. Enter this IP as the URL in the browser on your computer. Log in as the administrator by clicking on the “Admin login” button and following the instructions for a Static IP Connection.

Cisco SPA 112

How You Can Use the Phone Port Adapter:

1. Connect the AC adapter cord to the power connector and ethernet cable cord to the INTERNET connector on your phone.
2. Connect your analog phone to the adapter.
3. Dial “****” for voice menu activation.

Static IP connection:

1. Set the following parameters using the voice menu (according to documentation from the adapter):
 - “Enable\Disable DHCP” should be set to “Disabled”
 - “IP Address”
 - “Subnet”
 - “Gateway”
 - “DNS Server”

These parameters are provided by your ISP.

2. Return to the main menu and choose “Reset” to reboot your adapter.
3. Enter this IP as the URL in the browser on your computer. Log in as the administrator with “admin/admin” credentials.
4. Open the “Quick setup” tab and fill in the “Proxy”, “Display Name”, “Password”, “User ID” fields.

You can find your credentials here: Your Profile > My Forwarding numbers > Get connection details.

The screenshot shows the 'Quick Setup' page of the Cisco Phone Adapter Configuration Utility. The page is divided into two main sections: 'Line 1' and 'Line 2'. Each section contains fields for Proxy, Display Name, Password, Dial Plan, and User ID. Red circles with numbers 1 through 4 are overlaid on the form to indicate the fields mentioned in the text: 1 points to the Proxy field, 2 to the Display Name field, 3 to the Password field, and 4 to the User ID field.

5. Click the “Submit” button and choose the “Network setup” tab. Expand the “Basic Setup” section and select “Internet Settings”.
6. Set the “Static IP” value for the “Connection Type” field.
7. Fill “Internet IP Address”, “Subnet Mask”, “Default Gateway”, “Primary DNS” with values provided by your ISP. Click the “Submit” button.

Phone Adapter Configuration Utility

Quick Setup **Network Setup** Voice Administration Status

Basic Setup
Internet Settings
 Time Settings
 Advanced Settings

Internet Settings

Internet Connection Type

Connection Type: 1

Internet IP Address: . . . 2

Subnet Mask: . . . 3

Default Gateway: . . . 4

MTU:

Optional Settings

Host Name:

Domain Name:

DNS Server Order:

Primary DNS: . . . 5

Secondary DNS: . . .

8. Select the “Status” tab and check that your phone has a “Registered” status.

NAT connection:

1. Find out the IP address received by your phone adapter using the voice menu (according to documentation from the adapter).
2. Enter this IP as the URL in the browser on your computer. Log in as the administrator with “admin/admin” credentials.
3. Open the “Quick setup” tab and fill in the “Proxy”, “Display Name”, “Password”, “User ID” fields.

You can find your credentials here Your Profile > My Forwarding numbers > Get connection details.

Phone Adapter Configuration Utility

Quick Setup Network Setup Voice Administration Status

Quick Setup Quick Setup

Line 1	
Proxy:	prload.mightycall.com:16060
Display Name:	+18442071550
Password:	*****
Dial Plan:	(*xx[3469]110[00][2-9]xxxxxx1xxx[2-9]xxxxxx50)xxxxxxxxxxxxxx.)
Line 2	
Proxy:	
Display Name:	
Password:	
Dial Plan:	(*xx[3469]110[00][2-9]xxxxxx1xxx[2-9]xxxxxx50)xxxxxxxxxxxxxx.)

- Click the “Submit” button and choose the “Network setup” tab. Expand the “Basic Setup” section and select “Internet Settings”.
- Set the “Automatic Configuration - DHCP” value for the “Connection Type” field and click the “Submit” button.
- Select the “Status” tab and check that your phone has a “Registered” status.

Aastra 6731i

To get started, connect the AC adapter cord to the DC5V connector and the ethernet cable cord to the LAN connector on your phone.

- Press the “Options” button.





2. Move to the “Admin Menu” section and the enter default password: “22222”.
3. Choose “Network Settings”, then press “Enter”.
4. Select “DHCP Settings”, press “Enter”, and set the value of “Use DHCP?” to “Disabled”.
5. Return to “Network Settings” and select the “IP Address” section. Enter the IP address provided to you by your network provider.
6. Enter this IP as the URL in the browser on your computer. Log in as the administrator with “admin/22222” credentials.
7. Choose the “Advanced Settings” section, then select “Network”.
8. Fill in the “IP Address”, “Subnet Mask”, “Gateway”, “Primary DNS”, and “Secondary DNS” fields with values provided by your ISP.

Status System Information Operation User Password Phone Lock Programmable Keys Keypad Speed Dial Directory Reset Basic Settings Preferences Account Configuration Advanced Settings Network Global SIP Line 1 Line 2 Line 3 Line 4 Line 5 Line 6 Action URI Configuration Server Firmware Update TLS Support 802.1x Support Troubleshooting	<h2 style="margin: 0;">Network Settings</h2> <div style="background-color: #e6f2ff; padding: 2px; margin-bottom: 5px;">Basic Network Settings</div> <table style="width: 100%; border-collapse: collapse;"> <tr><td>DHCP</td><td style="text-align: right;"><input type="checkbox"/> Enabled</td></tr> <tr><td>IP Address</td><td style="text-align: right;"><input type="text" value="80.254.16.158"/></td></tr> <tr><td>Subnet Mask</td><td style="text-align: right;"><input type="text" value="255.255.255.252"/></td></tr> <tr><td>Gateway</td><td style="text-align: right;"><input type="text" value="80.254.16.157"/></td></tr> <tr><td>Primary DNS</td><td style="text-align: right;"><input type="text" value="80.254.17.2"/></td></tr> <tr><td>Secondary DNS</td><td style="text-align: right;"><input type="text" value="80.254.16.2"/></td></tr> <tr><td>Hostname</td><td style="text-align: right;"><input type="text" value="6731i00085D4E57B8"/></td></tr> <tr><td>LAN Port</td><td style="text-align: right;"><input type="text" value="Auto Negotiation"/></td></tr> <tr><td>PC Port PassThru Enable/Disable</td><td style="text-align: right;"><input checked="" type="checkbox"/> Enabled</td></tr> <tr><td>PC Port</td><td style="text-align: right;"><input type="text" value="Auto Negotiation"/></td></tr> </table> <div style="background-color: #e6f2ff; padding: 2px; margin-bottom: 5px;">Advanced Network Settings</div> <table style="width: 100%; border-collapse: collapse;"> <tr><td>DHCP Download Options</td><td style="text-align: right;"><input type="text" value="Any"/></td></tr> <tr><td>LLDP</td><td style="text-align: right;"><input checked="" type="checkbox"/> Enabled</td></tr> <tr><td>LLDP packet interval</td><td style="text-align: right;"><input type="text" value="30"/></td></tr> <tr><td>NAT IP</td><td style="text-align: right;"><input type="text" value="0.0.0.0"/></td></tr> <tr><td>NAT SIP Port</td><td style="text-align: right;"><input type="text" value="51620"/></td></tr> <tr><td>NAT RTP Port</td><td style="text-align: right;"><input type="text" value="51720"/></td></tr> <tr><td>STUN Server</td><td style="text-align: right;"><input type="text" value="0.0.0.0"/></td></tr> <tr><td>STUN Port</td><td style="text-align: right;"><input type="text" value="3478"/></td></tr> <tr><td>TURN Server</td><td style="text-align: right;"><input type="text" value="0.0.0.0"/></td></tr> <tr><td>TURN Port</td><td style="text-align: right;"><input type="text" value="3479"/></td></tr> <tr><td>TURN User ID</td><td style="text-align: right;"><input type="text"/></td></tr> <tr><td>TURN Password</td><td style="text-align: right;"><input type="text"/></td></tr> <tr><td>Rport (RFC 3581)</td><td style="text-align: right;"><input type="checkbox"/> Enabled</td></tr> </table> <div style="background-color: #e6f2ff; padding: 2px; margin-bottom: 5px;">HTTPS Settings</div> <table style="width: 100%; border-collapse: collapse;"> <tr><td>HTTPS Server - Redirect HTTP to HTTPS</td><td style="text-align: right;"><input type="checkbox"/> Enabled</td></tr> <tr><td>HTTPS Server - Block XML HTTP POSTs</td><td style="text-align: right;"><input type="checkbox"/> Enabled</td></tr> <tr><td>HTTPS Client Method</td><td style="text-align: right;"><input type="text" value="SSL 3.0"/></td></tr> <tr><td>Validate Certificates</td><td style="text-align: right;"><input checked="" type="checkbox"/> Enabled</td></tr> <tr><td>Check Certificate Expiration</td><td style="text-align: right;"><input checked="" type="checkbox"/> Enabled</td></tr> <tr><td>Check Certificate Hostnames</td><td style="text-align: right;"><input checked="" type="checkbox"/> Enabled</td></tr> <tr><td>Trusted Certificates Filename</td><td style="text-align: right;"><input type="text"/></td></tr> </table> <div style="background-color: #e6f2ff; padding: 2px; margin-bottom: 5px;">Type of Service DSCP</div> <table style="width: 100%; border-collapse: collapse;"> <tr><td>SIP</td><td style="text-align: right;"><input type="text" value="26"/></td></tr> <tr><td>RTP</td><td style="text-align: right;"><input type="text" value="46"/></td></tr> <tr><td>RTCP</td><td style="text-align: right;"><input type="text" value="46"/></td></tr> </table> <div style="background-color: #e6f2ff; padding: 2px;">VLAN</div>	DHCP	<input type="checkbox"/> Enabled	IP Address	<input type="text" value="80.254.16.158"/>	Subnet Mask	<input type="text" value="255.255.255.252"/>	Gateway	<input type="text" value="80.254.16.157"/>	Primary DNS	<input type="text" value="80.254.17.2"/>	Secondary DNS	<input type="text" value="80.254.16.2"/>	Hostname	<input type="text" value="6731i00085D4E57B8"/>	LAN Port	<input type="text" value="Auto Negotiation"/>	PC Port PassThru Enable/Disable	<input checked="" type="checkbox"/> Enabled	PC Port	<input type="text" value="Auto Negotiation"/>	DHCP Download Options	<input type="text" value="Any"/>	LLDP	<input checked="" type="checkbox"/> Enabled	LLDP packet interval	<input type="text" value="30"/>	NAT IP	<input type="text" value="0.0.0.0"/>	NAT SIP Port	<input type="text" value="51620"/>	NAT RTP Port	<input type="text" value="51720"/>	STUN Server	<input type="text" value="0.0.0.0"/>	STUN Port	<input type="text" value="3478"/>	TURN Server	<input type="text" value="0.0.0.0"/>	TURN Port	<input type="text" value="3479"/>	TURN User ID	<input type="text"/>	TURN Password	<input type="text"/>	Rport (RFC 3581)	<input type="checkbox"/> Enabled	HTTPS Server - Redirect HTTP to HTTPS	<input type="checkbox"/> Enabled	HTTPS Server - Block XML HTTP POSTs	<input type="checkbox"/> Enabled	HTTPS Client Method	<input type="text" value="SSL 3.0"/>	Validate Certificates	<input checked="" type="checkbox"/> Enabled	Check Certificate Expiration	<input checked="" type="checkbox"/> Enabled	Check Certificate Hostnames	<input checked="" type="checkbox"/> Enabled	Trusted Certificates Filename	<input type="text"/>	SIP	<input type="text" value="26"/>	RTP	<input type="text" value="46"/>	RTCP	<input type="text" value="46"/>
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RTP	<input type="text" value="46"/>																																																																		
RTCP	<input type="text" value="46"/>																																																																		

9. Go to the “Line 1” section.

10. At the “Basic SIP Authentication Settings” section, fill in “Screen Name”, “Screen Name 2”, “Caller ID”, “Authentication Name”, “Password” fields with values provided by your ISP.

Make sure that field “Phone Number” is equal to the “Authentication Name” value.

11. In the “Basic SIP Network Settings” section, fill in all “Server” and “Port” fields.

You can find your credentials here: Your Profile > My Forwarding numbers > Get connection details.

Set the value of the “Registration Period” field to “300” and make sure that the value labeled “Conference Server URI” is empty.

AASTRA

Status
System Information

Operation
User Password
Phone Lock
Programmable Keys
Keypad Speed Dial
Directory
Reset

Basic Settings
Preferences
Account Configuration

Advanced Settings
Network
Global SIP
Line 1
Line 2
Line 3
Line 4
Line 5
Line 6
Action URI
Configuration Server
Firmware Update
TLS Support
802.1x Support
Troubleshooting

Configuration Line 1

Basic SIP Authentication Settings

Screen Name: mc.softphones.1.gmail.c
Screen Name 2: mc.softphones.1.gmail.c
Phone Number: mc.softphones.1.gmail.c
Caller ID: 18442071550
Authentication Name: mc.softphones.1.gmail.c
Password: *****
BLA Number:
Line Mode: Generic
Call Waiting: Global

Basic SIP Network Settings

Proxy Server: prload.mytcall.com
Proxy Port: 16060
Backup Proxy Server: prload.mytcall.com
Backup Proxy Port: 16060
Outbound Proxy Server: prload.mytcall.com
Outbound Proxy Port: 16060
Backup Outbound Proxy Server: prload.mytcall.com
Backup Outbound Proxy Port: 16060
Registrar Server: prload.mytcall.com
Registrar Port: 16060
Backup Registrar Server: prload.mytcall.com
Backup Registrar Port: 16060
Registration Period: 300
Conference Server URI:

Advanced SIP Settings

AS-Feature-Event Subscription: Enabled
Park Pickup Config:

RTP Settings

DTMF Method: RTP
RTP Encryption: Global

Autodial Settings

Use Global Settings: Enabled
Autodial Number: -1
Autodial Timeout: 0

Save Settings

12. Go to the “System Information” section. The “Status” field should contain the “Registered” value.

Snom 300

How to configure your phone:

1. Connect the AC adapter cord to the power connector on your phone. Connect the ethernet cable cord to the NET connector on your phone.
2. Wait while your phone is fully loaded, then press the Menu button.

Static IP connection:

1. Find the section labeled “Configuration”, then select “DHCP”. Press the “X” button on the “Are you using DHCP?” dialogue.
2. Fill in the “IP Adr”, “Netmask”, “IP Gateway”, “DNS Server” fields with credentials provided by your internet provider. Your phone should then be rebooted.
3. Enter this IP address as the URL in the browser on your computer. Log in as the administrator with “admin/admin” credentials.
4. Select the “Setup” menu, choose “Identity 1”, and open “Login”.
5. Fill in the fields labeled “Account”, “Password”, “Registrar”, and “Authentication Username”. You can find your credentials here: Your Profile > My Forwarding numbers > Get connection details. Set the checkbox “Identity active” and click the “Save” button.


8
 VERSION

Configuration Identity 1

Operation

- Home
- Directory
- Setup**
- Preferences
- Speed Dial
- Function Keys
- Identity 1**
- Identity 2
- Identity 3
- Identity 4
- Action URL Settings
- Advanced
- Certificates
- Software Update

Status

- System Information
- Log
- SIP Trace
- DNS Cache
- Subscriptions
- PCAP Trace
- Memory
- Settings

Manual

[Login](#)
[SIP](#)
[NAT](#)
[RTP](#)

Login Information:

Identity active: on off ?

Displayname: ?

Account: ?

Password: ?

Registrar: ?

Outbound Proxy: ?

Failover Identity: None v ?

Authentication Username: ?

Mailbox: ?

Ringtone: Ringer 1 v ?

Custom Melody URL: ?

Display text for idle screen: ?

Ring After Delay (sec): ?

Record Missed Calls: on off ?

Record Dialed Calls: on off ?

Record Received Calls: on off ?

manual

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6. Move to the “SIP” tab and set the value of “Subscription Expiry” to “300”.

RTP/SAVP:	off	?
Media Transport Offer:	UDP	?
Media Transport Offer Setup:	active	?

7. Move to the “RTP” tab and set the value of the “Media Transport Offer” field to “UDP”.

RTP/SAVP:	off	?
Media Transport Offer:	UDP	?
Media Transport Offer Setup:	active	?

8. Click the “Save” button.

9. Move to the “Status” menu, select “System Information” and choose the “SIP Identity Status” section. Check that the value of “Identity 1 Status” is set to “Registered”.

NAT connection:

1. Find the section labeled “Information”, then find the “IP Adr” section and choose it. It should contain an IP address provided by your ISP.
2. Enter this IP address as the URL in the browser on your computer. Log in as the administrator with “admin/admin” credentials.
3. Select the “Setup” menu, choose “Identity 1” and open the “Login” tab.
4. Fill in the “Account”, “Password”, “Register”, “Authentication Username” fields.
You can find your credentials here: Your Profile > My Forwarding numbers > Get connection details.
Select the checkbox “Identity active” and click the “Save” button.

Configuration Identity 1

VERSION 8

Operation

Home

Directory

Setup

Preferences

Speed Dial

Function Keys

Identity 1

Identity 2

Identity 3

Identity 4

Action URL Settings

Advanced

Certificates

Software Update

Status

System Information

Log

SIP Trace

DNS Cache

Subscriptions

PCAP Trace

Memory

Settings

Manual



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[Login](#) [SIP](#) [NAT](#) [RTP](#)

Login Information:

Identity active: on off ?

Displayname: ?

Account: ?

Password: ?

Registrar: ?

Outbound Proxy: ?

Failover Identity: ?

Authentication Username: ?

Mailbox: ?

Ringtone: ?

Custom Melody URL: ?

Display text for idle screen: ?

Ring After Delay (sec): ?

Record Missed Calls: on off ?

Record Dialed Calls: on off ?

Record Received Calls: on off ?

5. Go to the "SIP" tab and set the value of "Subscription Expiry" to "300".

Remove all bindings on unregister: on off ?

Subscription Expiry (s): ?

Failed Subscription Retry Time (s): ?

Enable hook flash: on off ?

6. Go to the “RTP” tab and set the value of the “Media Transport Offer” field to “UDP”.

RTP/SAVP:	off	?
Media Transport Offer:	UDP	?
Media Transport Offer Setup:	active	?

7. Click the “Save” button.

8. Go to the “Status” menu, select “System Information” and choose “SIP Identity Status”. Check that the value of “Identity 1 Status” is set to “Registered”.

Yealink W52P

How to use and configure your phone:

Connect the AC adapter cord to the DC5V connector and the ethernet cable cord to the internet connector on your phone base station.

Static IP connection:

1. Press the “OK” button on your SIP phone.
2. Move to the System Settings menu, choose “Network” and enter your PIN with the value “0000”.
3. Set the value of “IP address Type” as Static. Fill “IPv4” with an IP address provided by your network provider.
4. Enter this IP address as the URL in the browser on your computer. Log in as the administrator with “admin/admin” credentials.
5. Choose the Network tab and select the Basic section.

Yealink W52P Logout

Status Account **Network** Phone Contacts Security

Basic **Advanced**

Internet Port

DHCP

Static IP Address

IP Address: 80.254.16.158

Subnet Mask: 255.255.255.252

Default Gateway: 80.254.16.157

Static DNS: On Off

NOTE

DHCP
The network configurations will be acquired from DHCP server.

Static IP Address
Specify the IP address, Subnet Mask, Default Gateway, Primary DNS, Secondary DNS fields manually.

Primary DNS 80.254.17.2

Secondary DNS 80.254.16.2

PPPoE
Contact your ISP if it should be used.

PPPoE

User

Password

Confirm Cancel

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6. Select the checkbox “Static IP Address” and fill the “IP address”, “Subnet Mask”, “Gateway”, “Primary DNS”, “Secondary DNS” fields with values provided by your internet provider.
7. Move to the Account tab and select the Register section. The Register Status field should contain a “Registered” value.

NAT connection:

1. Press the button labeled “OK” on your SIP phone.
2. Move to the “System Settings” menu, choose “Network” and fill the PIN field with a value of “0000”.
3. Set the “IPv4” value as DHCP.
4. Move to the “Status” section and choose “Base”. The field labeled “IP” should contain an IP address received by your phone.
5. Enter this IP address as the URL in the browser on your computer. Log in as the administrator with “admin/admin” credentials.
6. Choose the Account tab, then select the Advanced section.
7. Choose the Register tab. Set the “Line Active” field value to Enabled. Fill in the “Register Name”, “User Name”, “Password” fields. You can find your credentials here: Your Profile > My Forwarding numbers > Get connection details.
8. In the “SIP Server 1” field, fill the “Server Host”, “Transport”, “Server Expired” and “Port” fields with credentials provided by your SIP provider.
9. Click the Confirm button to save all changes.

Yealink | W52P Logout

Account Account1

Basic

Advanced

Codec

Number Assignment

Handset Name

Account

Register Status: Registered

Line Active: **1** Enabled

Label: **2** mc.softphones.1.gmail.com

Display Name: +18664908339

Register Name: **3** mc.softphones.1.gmail.com

User Name: **4** mc.softphones.1.gmail.com

Password: **5**

Enable Outbound Proxy Server: Disabled

Outbound Proxy Server: Port: 5060

Transport: **6** UDP

NAT Traversal: STUN

STUN Server: Port:

Proxy Require:

Anonymous Call: Disabled

Anonymous Call Rejection: Enabled

Get Balance: Disabled

Get Balance Link:

SIP Server 1

Server Host: **7** prioad.mytcall.com Port: **8** 16060

Server Expires: **9** 300

Server Retry Counts: 3

SIP Server 2

Server Host: Port: 5060

Server Expires: 3600

Server Retry Counts: 3

NOTE

Display Name
SIP service subscriber's name which will be used for Caller ID display.

Register Name
SIP service subscriber's ID used for authentication.

User Name
User account, provided by VoIP service provider.

NAT Traversal
Defines the STUN server will be active or not.

Proxy Require
A special parameter just for Nortel server. If you login to Nortel server, the value should be: com.nortelnetworks.firewall

Confirm Cancel

10. Select the Basic section. The Register Status field should contain a “Registered” value.

Yealink SIP-T19P

How to use and configure your phone:

Connect the AC adapter cord to the power connector on your phone. Connect the Ethernet

Connect the AC adapter cord to the power connector on your phone. Connect the Ethernet cable cord to the Internet connector on your phone.

Static IP connection:

1. Press the “Menu” button on the SIP phone.
2. Go to the Settings menu, choose “Advanced Settings”, and fill the “Password” field with “admin”.
3. Choose the Network section and select the “WAN Port”. Fill “IPv4” with the IP address provided by your network provider.
4. Enter this IP address as the URL in the web browser on your computer. Log in as the administrator with “admin/admin” credentials.
5. Choose the Network tab and select the Basic section.

The screenshot shows the Yealink T19_E2 web interface. The top navigation bar includes 'Status', 'Account', 'Network', 'DSSKey', 'Features', 'Settings', 'Directory', and 'Security'. The 'Network' tab is selected, and the 'Basic' section is active. The 'Internet Port' section is expanded, showing 'Mode(IPv4/IPv6)' set to 'IPv4'. Under 'IPv4 Config', the 'Static IP Address' radio button is selected. The IP Address field contains '80.254.16.158', Subnet Mask is '255.255.255.252', and Gateway is '80.254.16.157'. The 'Static DNS' section has 'On' selected, with Primary DNS '80.254.17.2' and Secondary DNS '80.254.16.2'. The 'IPv6 Config' section is also visible, with 'DHCP' selected. A 'NOTE' sidebar on the right contains information about DHCP, Static IP Address, PPPoE, and IPv6 Support.

6. Select the checkbox “Static IP Address” and fill in the “IP address”, “Subnet Mask”, “Gateway”,

“Primary DNS”, “Secondary DNS” fields with values provided by your internet provider.

7. Go to the Account tab and then select the Register section. The Register Status field should contain the “Registered” value.

NAT connection:

1. Press the “✓” button on your SIP phone to open the Status menu.
2. Check the “IPv4” field. It should contain IP-address, received by your phone.
3. Enter this IP address as the URL in the browser on your computer. Log in as the administrator with “admin/admin” credentials.
4. Choose the Register tab. Set the “Line Active” field value to Enabled. Fill in the “Register Name”, “User Name”, “Password” fields. You can find your credentials here: Your Profile > My Forwarding numbers > Get connection details.
5. In the “SIP Server 1” field fill the following fields:
 - “Server Host” - you can find your credentials here: Your Profile > My Forwarding numbers > Get connection details
 - “Transport” - UDP
 - “Server Expired” - 300
 - “Port” - 16060
6. Click Confirm to save all changes.
7. Move to the “Account” tab and select the Register section. The Register Status field should contain the “Registered” value.

Grandstream HT701

How to use your port adapter:

1. Connect the AC adapter cord to the power connector and the ethernet cable cord to the INTERNET connector on your phone.
2. Connect the analog phone to an adapter.
3. Dial “****” for voice menu activation.

Static IP connection:

1. Set the following parameters using the voice menu (according to documentation from your adapter):
 - a. “Enable\Disable DHCP” should be set to “Disabled”
 - i. “IP Address”
 - ii. “Subnet”
 - iii. “Gateway”
 - iv. “DNS Server”

These parameters are provided by your ISP.

2. Return to the main menu and choose “Reset” to reboot your adapter.

3. Enter this IP as the URL in the browser on your computer. Log in as the administrator with

- Enter this IP as the URL in the browser on your computer. Log in as the administrator with “admin/admin” credentials.
- Open the “FXS PORT” tab and set the “Account Active” value to “Yes”, and set the “Primary SIP Server” and “SIP Transport” values to “UDP”.
Fill in “SIP User ID”, “Authenticate ID”, “Authenticate Password”.
You can find your credentials here: Your Profile > My Forwarding numbers > Get connection details.

Grandstream Device Configuration

STATUS
BASIC SETTINGS
ADVANCED SETTINGS
FXS PORT

1 **Account Active:** No Yes

2 **Primary SIP Server:** (e.g., sip.mycompany.com, or IP address)
Failover SIP Server: (Optional, used when primary server no response)

Prefer Primary SIP Server: No Yes (yes - will register to Primary Server if Failover registration expires)

Outbound Proxy: (e.g., proxy.myprovider.com, or IP address, if any)

Allow DHCP Option 120(override SIP server): No Yes

3 **SIP Transport:** UDP TCP TLS (default is UDP)

NAT Traversal: No Keep-Alive STUN UPnP

4 **SIP User ID:** (the user part of an SIP address)

5 **Authenticate ID:** (can be identical to or different from SIP User ID)

6 **Authenticate Password:** (purposely not displayed for security protection)

Name: (optional, e.g., John Doe)

- Click the “Apply” button.
- Open the “STATUS” tab and check if the “Port Status” has a “Registered” value.

Grandstream Device Configuration

STATUS
BASIC SETTINGS
ADVANCED SETTINGS
FXS PORT

MAC Address: WAN-- 00:0B:82:95:19:2D (Device MAC)
IP Address: 80.254.16.158
Product Model: HT701
Hardware Version: V3.0B Part Number -- 9610001930B
Software Version: Program-- 1.0.8.2 Bootloader -- 1.0.0.9 Core -- 1.0.8.2 Base -- 1.0.8.2
 CPE --
System Up Time: 08:00:42 up 58 min
PPPoE Link Up: Disabled
NAT: Open Internet (STUN)

Port Status:	Port	Hook	User ID	Registration
	FXS	On Hook	mc.softphones.1.gmail.com	Registered

Port Options:	Port	DND	Forward	Busy Forward	Delayed Forward
	FXS	No			

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NAT connection

1. Find out the IP address using the Command 02 option in the voice menu.
2. Enter this IP address as the URL in the browser on your computer. Log in as the administrator with “admin/admin” credentials.
3. Open the “FXS PORT” tab and set the “Account Active” value to “Yes”, set the “Primary SIP Server” and “SIP Transport” values to “UDP”.

Fill in the “SIP User ID”, “Authenticate ID”, “Authenticate Password” values.

You can find your credentials here: Your Profile > My Forwarding numbers > Get connection details.

Grandstream Device Configuration

STATUS
BASIC SETTINGS
ADVANCED SETTINGS
FXS PORT

1 Account Active: No Yes

2 Primary SIP Server: (e.g., sip.mycompany.com, or IP address)
Failover SIP Server: (Optional, used when primary server no response)
Prefer Primary SIP Server: No Yes (yes - will register to Primary Server if Failover registration expires)
Outbound Proxy: (e.g., proxy.myprovider.com, or IP address, if any)

Allow DHCP Option 120(override SIP server): No Yes

3 SIP Transport: UDP TCP TLS (default is UDP)

4 NAT Traversal: No Keep-Alive STUN UPnP

5 SIP User ID: (the user part of an SIP address)

6 Authenticate ID: (can be identical to or different from SIP User ID)

7 Authenticate Password: (purposely not displayed for security protection)
Name: (optional, e.g., John Doe)

4. Open the “STATUS” tab and check if the “Port Status” has a “Registered” value.

Grandstream Device Configuration

STATUS
BASIC SETTINGS
ADVANCED SETTINGS
FXS PORT

MAC Address: WAN-- 00:0B:82:95:19:2D (Device MAC)

IP Address: 80.254.16.158

Product Model: HT701

Hardware Version: V3.0B Part Number -- 9610001930B

Software Version: Program-- 1.0.8.2 Bootloader -- 1.0.0.9 Core -- 1.0.8.2 Base -- 1.0.8.2
CPE --

System Up Time: 08:00:42 up 58 min

PPPoE Link Up: Disabled

NAT: Open Internet (STUN)

Port Status:

Port	Hook	User ID	Registration
FXS	On Hook	mc.softphones.1@gmail.com	Registered

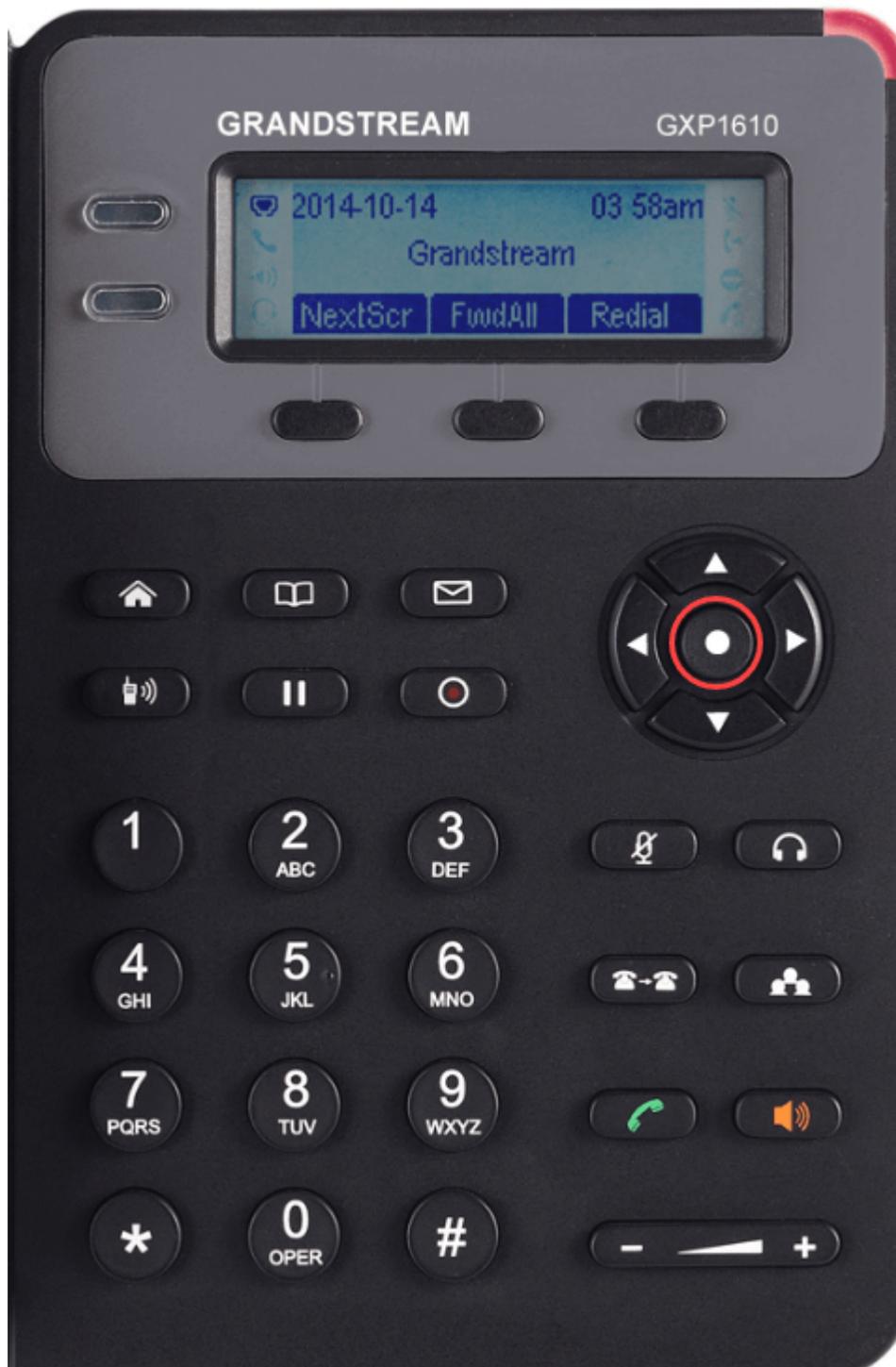
Port Options:

Port	DND	Forward	Busy Forward	Delayed Forward
FXS	No			

GrandStream GXP1610

How to configure your phone:

1. Connect the AC adapter cord to the power connector and the Ethernet cable cord to the INTERNET connector on your phone.
2. Wait until your phone is fully loaded, and then press the Menu button.



Static IP connection:

1. Open the “Network” section, go to “IP Settings” and choose the “Static IP” value.
2. Return to “WAN” settings and fill the “IP”, “Netmask”, “Gateway”, “DNS Server 1”, “DNS Server 2” fields with values provided by your ISP.
3. Return to the main menu and select the “Reboot” option.
4. Enter this IP as the URL in the browser on your computer. Log in as the administrator with “admin/admin” credentials.
5. Go to the “Accounts” tab, then select “Account 1” and choose the “General Settings” section.
6. Set the “YES” value for the “Account Active” field and fill in “Account Name”, “SIP Server”, “SIP User ID”, “Authenticate ID”.

You can find your credentials here: Your Profile > My Forwarding numbers > Get connection details.

Accounts

Account 1

General Settings

Network Settings

SIP Settings

Audio Settings

Call Settings

General Settings

Account Active	1 <input type="radio"/> No <input checked="" type="radio"/> Yes
Account Name	2 <input type="text" value="mc.softphones.1.gmail.c"/>
SIP Server	3 <input type="text" value="pload.mycall.com:1606"/>
Secondary SIP Server	<input type="text"/>
Outbound Proxy	<input type="text"/>
Backup Outbound Proxy	<input type="text"/>
SIP User ID	4 <input type="text" value="mc.softphones.1.gmail.c"/>
Authenticate ID	5 <input type="text" value="mc.softphones.1.gmail.c"/>
Authenticate Password	<input type="text"/>
Name	<input type="text"/>
Voice Mail Access Number	<input type="text"/>

Save
Save and Apply
Reset

7. Click the “Save” button.

8. Open the “SIP Settings” and select the “Basic Settings” section. Set the “SIP Registration” field

8. Open the SIP Settings and select the Basic Settings section. Set the SIP Registration field to “Yes”, set “5” as the value for “Register Expiration”, “Instance” for “Unregister on Reboot” and “UDP” for the “SIP Transport” field.

Accounts

Account 1

General Settings

Network Settings

SIP Settings

Basic Settings

Advanced Features

Session Timer

Security Settings

Audio Settings

Call Settings

Basic Settings

TEL URI Disabled User=phone Enabled

SIP Registration 1 No Yes

Unregister on Reboot 2 No All Instance

Register Expiration 3

Reregister before Expiration

Enable OPTIONS Keep Alive No Yes

OPTIONS Keep Alive Interval

OPTIONS Keep Alive Max Lost

Local SIP Port

SIP Registration Failure Retry Wait Time

SIP T1 Timeout

SIP T2 Timeout

SIP Transport 4 UDP TCP TLS/TCP

9. Click on the “Save and Apply” button.

10. Go to the “Status” tab and select “Account Status”. There should be a “YES” in front of your SIP account.

Grandstream GXP1610
Admin Logout | Reboot | Factory Reset
English

GRANDSTREAM

CONNECTING THE WORLD

STATUS

ACCOUNTS

SETTINGS

NETWORK

MAINTENANCE

PHONEBOOK

Home
Version 1.0.2.27

Status

Account Status

Account Status		Account	SIP User ID	SIP Server	SIP Registration
Network Status		Account 1	mc.softphones.1.gmail.com	prload.mytcall.com:16060	YES
System Info					

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NAT connection

1. Open the “Network Status” section and check for the IP in the IPv4 field.
2. Enter this IP as the URL in the browser on your computer. Log in as the administrator with “admin/admin” credentials.
3. Move to the “Accounts” tab, then select “Account 1” and choose the “General Settings” section.
4. Set a “YES” value in the “Account Active” field, and fill in the “Account Name”, “SIP Server”, “SIP User ID”, “Authenticate ID” fields.

You can find your credentials here: Your Profile > My Forwarding numbers > Get connection details.

Accounts

Account 1

General Settings

Network Settings

SIP Settings

Audio Settings

Call Settings

General Settings

Account Active 1 No Yes

Account Name 2

SIP Server 3

Secondary SIP Server

Outbound Proxy

Backup Outbound Proxy

SIP User ID 4

Authenticate ID 5

Authenticate Password

Name

Voice Mail Access Number

5. Click the “Save” button.
6. Open the “SIP Settings” tab and select “Basic Settings”. Set “Yes” for the “SIP Registration” field, set a “5” value for “Register Expiration”, “Instance” for “Unregister on Reboot” and a “UDP” value for the “SIP Transport” field.

Accounts

Account 1

General Settings

Network Settings

SIP Settings

Basic Settings

Advanced Features

Session Timer

Security Settings

Audio Settings

Call Settings

Basic Settings

TEL URI Disabled User=phone Enabled

SIP Registration 1 No Yes

Unregister on Reboot 2 No All Instance

Register Expiration 3

Reregister before Expiration

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OPTIONS Keep Alive Interval

OPTIONS Keep Alive Max Lost

Local SIP Port

SIP Registration Failure Retry Wait Time

SIP T1 Timeout

SIP T2 Timeout

SIP Transport 4 UDP TCP TLS/TCP

7. Click the “Save and Apply” button.
8. Move to the “Status” tab and select “Account Status”. There should be a “YES” in front of your SIP account.

Grandstream GXP1610
Admin Logout | Reboot | Factory Reset
English

GRANDSTREAM

CONNECTING THE WORLD

STATUS
ACCOUNTS
SETTINGS
NETWORK
MAINTENANCE
PHONEBOOK

Home
Version 1.0.2.27

Status

Account Status

Account Status			
Account	SIP User ID	SIP Server	SIP Registration
Account 1	mc.softphones.1@gmail.com	prload.mytcall.com:16060	YES

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Using Telephony User Interface (TUI) On Your Deskphone

How to make a direct transfer:

Press ****101#** to make a direct transfer.

“101” is an extension of a team member in your MightyCall account. You can use any phone number, for example, 18882568312 as well.

How to make a Consultative Transfer:

Press **##** sequentially during a conversation and wait for an audio guide to play. The following options are available:

Press **1** — make a call to another person (a second line).

Press **2** — merge all active calls into a conference.

Press **3** — disconnect the second line.

Press **4** — return to the first line (the second line will be put on hold).

For example:

During a conversation press:

101 # [call another person (second line) ... conversation with the second line] **## 1** or hang up — the initial call will be transferred.

101 # [call another person (second line) ... conversation with the second line] **## 2** — a conference call for 3 members.

101 # [call another person (second line) ... conversation with the second line] **## 3** — transfer the initial call to another party.

101 # [call another person (second line) ... conversation with the second line] **## 4** — return to the first line, the second line will be put on hold.

Did this answer your question?



