

# Yealink@Home

## **Requirements and Configuration**

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## 1 Yealink@Home

You can connect a Yealink desk phone to SwyxWare from any location outside a virtual private network.



In this scenario, you can use a Yealink desk phone only as a SIP terminal with basic IP telephony features.

## 1.1 Concept

A SwyxWare user can connect his Yealink desk phone at home to a DSL router with Internet access, log on to SwyxServer with his SIP credentials and use the SwyxWare telephony functionality.



It is ensured that SIP messages are only forwarded by registered users.

#### **DSL** gateway

The static public IP address of the DSL gateway serves as the SIP registrar, SIP proxy, and media endpoint for the Yealink phones. Using appropriate port forwarding rules, the received encrypted SIP and RTP packets are forwarded to the AudioCodes Session Border Controller (SBC).

#### AudioCodes SBC

AudioCodes' Session Border Controller receives the encrypted SIP and RTP packets, decrypts them and forwards them to SwyxWare.



Abbildung 1: Netzwerkübersicht, SwyxWare

### 1.2 **Requirements**

#### SwyxWare

- Installed and configured SwyxWare environment
- On SwyxWare a SIP trunk with the corresponding public phone numbers should be set up, see https://help.swyx.com/cpe/latest.version/Administration/Swyx/en-US/ index.html#context/help/sip\_connections\_\$
- Public phone numbers should be assigned to SwyxWare users, see https://help.swyx.com/cpe/latest.version/Administration/Swyx/en-US/ index.html#context/help/assign\_numbers\_\$
- To operate a Yealink desk phone as a pure SIP end device from the home office, SIP credentials have to be configured for the respective user in SwyxWare Administration, see

https://help.swyx.com/cpe/latest.version/Administration/Swyx/en-US/ index.html#context/help/SIP\_registry\_\$

#### AudioCodes SBC

- Licensed AudioCodes Session Border Controller (1x FEU license per registration)
- AudioCodes Firmware v7.20 and higher



You can change the ports or port ranges if necessary. You must then take the corresponding entries into account when configuring the SBC and the Yealink desk phones.

• AudioCodes SBC must be preconfigured for use with Yealink@Home, see *Configure AudioCodes SBC*, page 3

#### Edge / DSL Gateway

- Static, public IP address for the DSL gateway
- On the DSL gateway, the following port forwarding must be set up from the public ports to the AC SBC:
  - SIP Signaling Port (TLS): 2606
  - Media Ports: e.g 8000 8199

#### Yealink

- Yealink desk phone (T41S, T42S, T46S, T48S)
- Yealink desk phone must be updated to the latest firmware, at least to version 66.84.133.2

## **1.3 Configure AudioCodes SBC**

The AudioCodes Session Border Controller can be configured using the commented CLI file.



You can change the ports or port ranges if necessary. You must then take the corresponding entries into account when configuring the SBC and the Yealink desk phones.

In addition, if the ports are changed, the DSL gateway must also be adjusted accordingly.

#### To import the CLI file with your configuration data to the SBC

- 1 Download the ZIP file from our web page: https://www.swyx.de/ download.php?filen=Yealink\_at\_HOME\_M500L\_LAN.zip&loc=extern
- 2 Unzip the CLI file into a folder and open it in a text editor.
- 3 Adjust the configuration data according to the pertaining comments and save your changes.
- 4 Open the AudioCodes web interface.
- 5 In the upper menu bar, open "Actions | Configuration File" on the right.

SET	UP	MONITOR	TROUBLESHO	т	Save	Reset	Actions <del>-</del>	4	Admin <del>-</del>
G & N	MEDIA		TRATION				Configurat	ion File	alue
							Auxiliary F	iles	
							License Ke	у	
C	Config	guration File					Software L	Ipgrade	
								10.111	_
	INI FI	ILE			RESTORE THE I	DEFAULT CONF	IGURATION OF 1	THE DEVICE	
	Save	e INI file to the PC		Save INI File		Restore Facto	ory Defaults		
	Load	d <b>INI</b> file to the de	vice.						
	Cho	oose File No file	chosen	Load INI File					
	CLI S	SCRIPT							
	Save	CLI Script file to	the PC	Save CLI Script File					
	Load	d CLI Script file to	the device.	Sole ensempline					
	Cho	oose File No file	chosen	Load CLI Script File					
	Load	d CLI Startup Scrip	ot to the device.						
	Cho	oose File No file	chosen	Load CLI Startup Script					
	CON	IFIGURATION PAC	KAGE						
	Save	Configuration Pa	ckage to the PC	Save Configuration Package					
	Load Co	onfiguration Pack	age to the device.	ave computation rackage					
	Choos	e File No file ch	osen	Load Configuration Package					

6 Under "Load CLI Startup Script to the device" click on "Choose File" and select the CLI file.



The existing configuration on the SBC will to be completely overwritten by the data in the CLI file.

- 7 Click on "Load CLI Startup Script".
  - ✓ Your configuration data has been transferred to the device. The process may take a few minutes. The Session Border Controller can be restarted 1 or 2 times.

#### **Configuration of the desk phone** 1.4

For configuration use the web interface from the manufacturer.

If you would like to delete any existing configuration, perform a factory reset.

You have to make the following settings under "Account | Register":

Yealink Tase	Default password is in use. Please changel English (English)							
	Status Account Netwo	rk Dsskey Fe	atures Set	tings	Directory	Security		
Register	Account	Account1	• 0		NOTE			
Register	Register Status	Registered			Account Reg	istration		
Basic	Line Active	Enabled	• 🕜		Register accou phone.	int (s) for the IP		
Codec	Label	User3622	0		Server Redu	ndancy		
Advanced	Display Name	User3622	0		It is often requ development t	ired in VoIP o ensure service		
	Register Name	3622	0		continuity, for server needs t	events where the o be taken offline		
	Username	3622	0		for maintenan when the conr	ce, or for events lection between		
	Password	•••••	0		the IP phone a	ind the server fails		
	SIP Server 1 🕜				A computer ne	tworking techniqu		
	Server Host	80.153.207.2	Port 2606	0	Internet proto	col connections		
	Transport	TLS	. 0		NAT.	fo chae impreniene		
	Server Expires	120	0		You can config	jure NAT traversal		
	Server Retry Counts	3	0		for this accour	it.		
	SIP Server 2 🕜				Click here	e to get more		
	Server Host		Port 5060	0	product docu	ments.		
	Transport	UDP	• 🕜					
	Server Expires	3600	0					
	Server Retry Counts	3	0					
	Enable Outbound Proxy Server	Disabled	• 🕜					
	Outbound Proxy Server 1		Port 5060	0				
	Outbound Proxy Server 2		Port 5060	0				
	Proxy Fallback Interval	3600	0					
	NAT	Disabled	. 0					

Designation	Explanation			
Line Active	Select "Enabled" to activate the line.			
Label	Enter the display name of the SwyxWare user			
Display Name	under "Label" and "Display Name" respectively (This information may differ from the SwyxWare configuration).			
Register Name	Enter the SIP credentials of the SwyxWare user			
Username	that you have configured in the SwyxWare Administration			
Password				

Designation	Explanation
Server host	Enter the public IP address and the port of the DSL gateway.
Transport	Select "TLS".
Server Expires	Default value: 120

• The following setting must be made under "Account | Advanced":

Vaalink				Logout English (English) 🔻
	Status Account Network	c Dsskey Feature	es Settings	Directory Security
Register	Account	Account1 🔻	0	NOTE
Register	Keep Alive Type	Default 🔹	0	DTMF
Basic	Keep Alive Interval (Seconds)	30	0	It is the signal sent from the IP phone to the network, which is
Codec	RPort	Disabled 🔹	0	generated when pressing the IP phone's keypad during a call.
Advanced	Subscription Period (Seconds)	1800	0	
	DTMF Type	SIP INFO 🔻	0	Session Timer It allows multiple participants
	DTMF Info Type	DTMF-Relay 🔻	0	(more than three) to join a call.
	DTMF Payload Type (96~127)	101	0	VQ-RTCPXR The VO-RTCPVR mechanism
	Retransmission	Disabled 🔻	0	complaint with RFC 6035, sends
	Subscribe Register	Disabled 🔻	0	contained SIP PUBLISH messages to the central report collector.
	Subscribe for MWI	Enabled 🔻	0	
	MWI Subscription Period (Seconds)	3600	0	Click here to get more
	Subscribe MWI to Voice Mail	Disabled 🔻	0	produce documents.
	Voice Mail	##10	0	
	Voice Mail Display	Enabled 🔻	0	
	Caller ID Source	FROM	0	
	Session Timer	Enabled 🔻	0	
	Session Expires (90~7200s)	90	0	
	Session Refresher	UAC 🔻	0	
	Send user=phone	Enabled 🔹	0	
	RTP Encryption (SRTP)	Compulsory •	0	
	PTime (ms)	20 🔻	0	
	BLF List URI		0	
	BLF List Pickup Code		0	
	BLF List Barge in Code		0	
	BLF List Call Parked Code		0	

Activate the voice encryption: Set the "RTP Encryption (SRTP)" option to "Compulsary" (required).

			Logout
Vealink		A Default password is	in use. Please change! English (English) 🗸
	Status Account Network	Dsskey Features Setti	ngs Directory Security
Preference	Time&Date		NOTE
Time&Date	DHCP Time Manual Time	Disabled V ?	Time and Date It displays on the idle screen of
Call Display	Time Zone	+1 Spain、France、Germany、Italy、Netherlan	<ul> <li>Time Zone</li> </ul>
Upgrade	Daylight Saving Time	● Automatic ○ Enabled ○ Disabled ?	A time zone is a region on Earth that has a uniform standard time.
Auto Provision	Location	Germany(Berlin) V	commercial or other communication to keep the same
Configuration	Fixed Type Start Date	DST by Date DST by Week March V Last Week V Sunday V 02:00	V NTP Server
Dial Plan	End Date	October 🗸 Last Week 🗸 Sunday 🖌 03:00	<ul> <li>The IP phones synchronize the time and date automatically from the NTP time server by default</li> </ul>
Voice	Offset (minutes)	60	Daylight Saving Time
Ring	NTP by DHCP Priority	High 🗸 🕜	It is the practice of temporary advancing clocks during the
Tanaa	Primary Server	cn.pool.ntp.org	have more daylight and mornings have less. Typically, clocks are
Tones	Secondary Server	pool.ntp.org	adjusted forward one hour at the start of spring and backward in
Softkey Layout	Update Interval (15~86400s)	1000 V	autumn.
TR069	Date Format	WWW MMM DD V	Click here to get more
Voice Monitoring			produce documents.
SIP	Confirm	Cancei	
Power Saving			

Set the correct time zone, location, and daylight saving time rule.

• The following setting must be made under "Security | Trusted Certificates":

#### • The following setting must be made under "Setting | Time&Date":

Vealink							English (Engli	Logout ish) <b>v</b>		
	Status	Account	Network	c Dsskey	Features	Settings	Directory	urity		
Password	Index Id	Issued to		Issued by	Expiration	Delete	NOTE	•		
Trusted Certificates	1						Transport Layer Sect (TLS) Trusted Certificate	urity		
Server Certificates	2						When the IP phone requests a TLS connection with a server,			
	3						the IP phone should ve certificate sent by the s	erify the server to		
	4						decide whether it is tru based on the trusted ce	ertificates		
	5						trusted certificates. You	u can		
	6						most. The format of the truste			
	7						*.pem,*.cer,*.crt and * the maximum file size i	der and is 5MB.		
	8									
	9						Click here to get more product documents.			
	10									
						Delete	]			
	Or	ly Accept Trusted Ce	rtificates	Disabled	v 📀					
	Co	mmon Name Validati	on	Disabled	· (2)					
	CA	Certificates		All Certificates	• 0					
	Impo	rt Trusted Certifica	tes 🕜							
	Lo	ad Trusted Certificate	s File	No selected file	Browse					
				Upload	load Cancel					
		Γ	Confirm	Cai	ncel					

The phone must allow TLS connections without certificate checking. Set the "Only Accept Trusted Certificates" option to "Disabled".

## 1.5 Verification

After successful configuration, you can check the login status of the Yealink desk phone connected to SwyxWare via Yealink@Home in the Swyx-Ware Administration.

In addition, the SIP messages can be monitored using AudioCodes' Syslog Viewer:

http://redirect.audiocodes.com/install/index.html