T300/T500 - Grandstream HT-702/704 ATA - How To





This article will describe the configuration of the HandyTone 702/704 behind the Gigaset T300/500 PBX system.

To guarantee a proper operation, please update the device first to the latest sw-version (V1.0.4.8, 05.02.2013).

The automatic (one-touch) setup is supported with the HT702 and HT704(PBX version 5.3.x.x or higher). The HT701 must be configured manually. The manual steps show the HT701.

ATTENTION

Only setup for this device is supported by Gigaset.

No further support from Gigaset in operational state for this device !!!

This section describes the manual setup of the HT701, HT702 and HT704 behind the Gigaset T300/500 PBX system.

For the automatic setup please scroll down.

In order to use a Grandstream VoIP ATA to connect a fax machine or analogue telephone to the T300/T500 PBX you have to take the following steps:

- Connect the ATA Adapter to your network and connect the powercable. To get the IP address of the HT70X connect an analog phone to the line connector of the box and dial ***. The voice menu asks you to choose the wanted menu. Dial 02. The voice system will tell you the current IP address of the box.
- 2. Open your webbrowser and type in the address, e. g. 192.168.2.24. The default password of the box is admin.
- 3. The basic and advanced settings can be kept by default.
- 4. The FXS Port Menu has to be configured as shown.

Grandstream Device Configuration				
STATUS BASI	IC SETTINGS ADVANCED	SETTINGS FXS PORT		
Account Active:	💿 No 💿 Yes			
Primary SIP Server:	192.168.2.50	(e.g., sip.mycompany.com, or IP address)		
Failover SIP Server:		(Optional, used when primary server no response)		
Prefer Primary SIP Server:	No O Yes (yes - with a second seco	Il register to Primary Server if Failover registration expires)		
Outbound Proxy:	192.168.2.50	(e.g., proxy.myprovider.com, or IP address, if any)		
SIP Transport:	O UDP O TCP O TLS (default is UDP)			
NAT Traversal (STUN):	No O No, but send kee	ep-alive 🔘 Yes		
SIP User ID:	GrandHT701	(the user part of an SIP address)		
Authenticate ID:	GrandHT701	(can be identical to or different from SIP User ID)		
Authenticate Password:		(purposely not displayed for security protection)		
Name:	HT701	(optional, e.g., John Doe)		
DNS Mode:	: • A Record O SRV O NAPTR/SRV			
Tel URI:	Disabled 🔻			
SIP Registration:	No Ves Yes Yes			
Unregister On Reboot:	No O Yes			
Outgoing Call without Registration:	O No O Yes			
Register Expiration:	60 (in minutes. default 1 hour, max 45 days)			
SIP Registration Failure Retry Wait Time:	20 (in seconds. Between 1-3600, default is 20)			
Local SIP port:	5060 (default is 5060 for UDP and TCP; 5061 for TLS)			
Local RTP port:	5004 (1024-65535, default 5004)			

In our example the T500 PBX uses 192.168.2.50 as IP address. The SIP name configured in the PBX is GrandHT701, we use a random password. Gigaset PRO Administration - Mozilla Firefox

192.168.2.50/config/display.do?url=undefined					
Configuration	X				
	Edit				
Users	Details				
Groups	Telephone Type: Standard Sip Y Functionality: Phone Y				
Phones					
🖗 Modules	Applied server address: 192.168.2.50 Last known Device-IP: 192.168.2.24				
Voicemail	Autoprovisioning				
Conference	Last provisioning: lan 1 1970 12:00:00 AM				
Addressbook	MAC address:				
71 ² Phone Numbers	Firmware version:				
Lines	Extended Configuration				
Routing	NAT: default 🗸				
C Server	Codecs: alaw,ulaw				
Statistic	Restrict to IP:				
	Door Intercom				
Security	Canable				
*+ Extended Settings	Camera URL: Test Connection DTMF:				
	Save Apply Cancel				

We advice to check the codec settings at the bottom of the FXS port menu. We strongly advice only to use the G.711 alaw /ulaw codec for use with connected fax machines.

Use First Matching Vocoder in 2000K SDP:	No O Yes
Preferred Vocoder:	choice 1: PCMA -
(in listed order)	choice 2: PCMU -
	choice 3: PCMA -
	choice 4: PCMA -
	choice 5: PCMA -
	choice 6: PCMA
G723 Rate:	6.3kbps encoding rate 5.3kbps encoding rate
iLBC Frame Size:	20ms 0 30ms 30ms
iLBC Payload Type:	97 (between 96 and 127, default is 97)
VAD:	No Ves
Symmetric RTP:	No Ves
Fax Mode:	I.38 Pass-Through
Re-INVITE After Fax Tone Detected:	Inabled O Disabled
Jitter Buffer Type:	© Fixed
Jitter Buffer Length:	💿 Low 💿 Medium 💿 High
SRTP Mode:	Disabled Enabled but not forced Enabled and forced
SLIC Setting:	GERMANY
Caller ID Scheme:	Bellcore/Telcordia -

This section describes the automatic setup of the HT702 and HT704 behind the Gigaset T300/500 PBX system.

In order to use the one-touch provisioning, please open the admin-section of the PBX and go to *Phon* es. Start here the phonesearch by pressing the '**Se arch**'-Button in the right-lower corner

Con								
		List of configured phones						
	Users	Settings	Configured Phones	Extended Settings				
ĉ	Groups			aarch	Dours: 10 st		1/2	
T	Phones			saten	Rows. 10 V		112	
Ø	Modules	State	Device Type	Device Name	IP	Assigned Persons		
-	Voicemail		DE310 IP PRO	1081.DE310IP	192.168.16.133	Lahm, Phillip	6	×
60	Conference		DE410 IP PRO	10/9.DE410IP	192.168.16.131	Schweinsteiger, Bastian	6	X
Ð	Addressbook		DE700 IP PRO	1151.DE/00IP	192.168.16.195	Oezil, Mesut	6	×
7,2	Phone Numbers		DE900 IP PRO	1080.DE900IP	192.168.16.132	Neuer, Manuel	4	X
	Lines		N510 IP PRO	1117.N510IP	192.168.16.194	Klose, Miroslav	6	×
-	Dautian		N510 IP PRO	1119.N510IP	192.168.16.194	Mueller, Thomas	6	X
	Routing		N510 IP PRO	1118.N510IP	192.168.16.194		6	×
	Server		N510 IP PRO	1122.N510IP	192.168.16.194		6	X
\sim	Statistic		N510 IP PRO	1120.N510IP	192.168.16.194		6	×
品	Interconnection		N510 IP PRO	1121.N510IP	192.168.16.194		6	×
8	Security							+
*+	Extended Settings							
]						
		Add Additi	onal Telephones:	Manually				Search
							[Close

You can now specify the IP-range you want to search for the device or just start the search. When the device is found it will be automatically provisioned with the SIP-account.

After this you can start assigning the accounts to the users.

Additional Information:

External Link

Configuration			X	
	Automatic Scan for Telephones			
Users	Search now	Applied server address:	192.168.16.191	
Groups	0%	IP range to search:	192.168.16.0 - 255	
Phones	Configured Devices			
Modules	Quantity Type			
Voicemail				
Conference				
Addressbook				
71 ² Phone Numbers				
Lines				
Routing				
Server				
Statistic				
Security				
*+ Extended Settings				
	Add Additional Telephones: Manually		Specify IP range to search	
			Close	