

Interop T440 / T640 - VTX Telecom (FW Version 1.1.0)

VTX Switzerland

Feature	
Outgoing Calls	Yes
Incoming Calls	Yes
CLIP incoming	Yes
CLIP outgoing	Yes
Call Forwarding	Yes
Call Transfer	Yes
Call Waiting	Yes
DTMF	Yes
Anonymous Call	n/a
A-number forwarding	Yes

- [VTX Switzerland](#)
 - [Gigaset T440/T640 PRO settings.](#)
 - [Gateway Group](#)
 - [SIP gateway](#)
 - [Inbound routes](#)
 - [Outbound routes](#)



Gigaset T440/T640 PRO settings.

For entering a new SIP trunk into the PBX, you need some steps:

1. Adding a new Gateway group
2. Adding a new SIP gateway
3. Defining the inbound routes (assignment of number to extension)
4. Defining the outbound routes

Gateway Group

In the Gigaset PBX go to "Administration" - "Routes" - "Gateway groups" enter a name for the new group and click on: **Create new group**

In the Gateway group you have to define the Outbound Caller ID, the Inbound DIDs (how the number is forwarded to the Inbound routes) and the Inbound caller ID (number presentation of external caller). in addition you can permit here inbound calls in general for this gateway group.

Home
Menu
Administration

Profile

Users & extensions >
 System >
 Provisioning >
 Routes >
Gateway groups
 SIP gateways
 TDM Gateways
 Inbound routes
 Outbound routes
 Call forward
 System status >

Title

VTX

Permit inbound calls

ON

Outbound caller ID

Search/replace pattern for outbound caller ID (1)
 s/ / /

Asserted Identity

Search/replace pattern for asserted identity (1)
 s/ / /

Inbound DIDs

Search/replace pattern to cut prefixes (2)
 s/ / /

Inbound caller ID

Search/replace pattern for inbound caller ID (3)
 s/ / /

Gateways

SIP - VTX

(1) Search/Replace pattern (PCRE) for outbound caller ID signalling. Examples:
 Extension only: s/^(*)/\$1/
 National format: s/^(\d{3})030123456\$1/
 International format: s/^(\d{3})004930123456\$1/ or s/^(\d{3})+4930123456\$1/
 Same number for all users: s/^(\d{3})00493012345612/
 In most cases you should use the national or international format.

(2) If necessary, specify here a PCRE-pattern which removes prefixes from inbound numbers, so that only the internal extension remains. Examples:
 s/^026313370//
 s/^((00490)2631|03370)/
 s/^(\d{3}00490)2631|03370/.*\$1/

(3) Search/Replace pattern (PCRE) for inbound caller IDs, to fix up eventually wrong signalled caller IDs. Example: s/^0/ or s/^(\d{3})0\$1/ to prefix the caller ID with a 0. Empty for no automatic change.
 Experimental: This option can also be used to add or remove a leading 0 to/from call log entries by using s/^0/ or s/^(\d{3})0\$1/ or s/^0/ or s/^0(\d{3})\$1/.

Outgoing caller IDs

Extension	Caller ID	
101	0325661425	
102	0325661426	
Extension <input type="text"/>	Caller ID <input type="text"/>	

Cancel
Save

Outbound caller ID

As the number is fixed with this account, every user who is allowed to use this line will present the same number to the callee.

Asserted Identity

These fields are used specially for external forwardings. Some providers need special settings in order to present the correct number at the receiving party.

Inbound DIDs

For incoming calls, you can use the regular expression, generated by the setup assistant. In your case you would have to adjust the city-code and the first part of the extension, according to your line-settings.

Example: `^(?:0041|+41|3210)7567032566142(?:*)`

This will be forwarded to the Inbound Routes.

Inbound caller ID

For incoming calls, no additional **0** will be added.

Gateways

This field will be empty when you create this Gateway group. It will show later the assigned SIP gateways.

SIP gateway

In the Gigaset PBX go to "Administration" - "Routes" - "SIP gateways" enter a name for the new gateway and click on: **Create new gateway**

The SIP gateway contains all necessary data for the registration and dial command and how the number is delivered to/from the provider.

Just enter or choose the values according to your contract or data you received from the provider.

Example of VTX gateway settings 1/2.

Home Menu Administration
Gigaset Profile

- Users & extensions >
- System >
- Provisioning >
- Routes >
 - Gateway groups
 - SIP gateways**
 - TDM Gateways
 - Inbound routes
 - Outbound routes
 - Call forward
 - System status >

Title	VTX
Name	gw_1_vtx
Registrar	s1.259634.trk.ipvoip.ch
Proxy [1]	s1.259634.trk.ipvoip.ch
User [2]	413
Password	
Allow outbound calls	<input checked="" type="checkbox"/>
Register	<input checked="" type="checkbox"/>
Language	en - English (en-US)
Dial comand [3]	SIP/(number)@(gateway)
Source of destination number	INVITE request line
Group [4]	VTX
Port [5]	5060
NAT	yes
Redirect RTP stream	Do not reroute RTP stream (default)
Check availability	<input checked="" type="checkbox"/>
Simultaneous calls	0 0 for unlimited (default)
DTFM mode	rtf2833 - RTP meta data
From user	
From Domain	
T38 support	<input type="checkbox"/>

(1) Empty for no proxy.

(2) For some SIP providers, it might be necessary to use the format user@domain. (domain is then used in the From header, which equals fromdomain in Asterisk.)

(3) String for the Dial() command. T440/T640 PRD will automatically replace (number) by the called number, (number1) without the first digit and (gateway) with the internal description of gw_1_ami.

(4) In order to use gateways, they must be assigned to a gateway group.

(5) When specifying the port (standard SIP port: 5060), it will be directly used. Without, a DNS lookup of the SRV record _sip._udp of the domain (or server) will be performed upon dialout. More Information [Srv Resource Record](#) (en), [CIDR](#), [SIP-DNS-Srv-Records](#), [SIP-DNS-Srv-Records](#) (en), [SIP-DNS-Srv-Records](#) (en)

(6) Useful settings are e.g.
0.0.0.0/0 to allow calls from all IP addresses.
192.0.2.0/24 to allow calls originating from network 192.0.2.*.
192.168.0.0/16 to allow calls originating from network 192.168.*.*.
192.168.1.1/32 to allow calls originating from IP address 192.168.1.1 etc.
More Information [IP address](#), [Subnet](#), [CIDR](#), [sip.conf](#)

(7) The priority of the codes is from left to right and top to bottom

Dial command

The dial command is the command which is used in the asterisk software. The term **{number:1}** means, that at the dialled number (e.g. 008987654321) the first digit is removed. If you don't use a line access code (in most cases '0' is used), you have to remove the ':1' !!! (Example is without access code)

Group

Select here the previously created gateway group.

Update remote party ID (CLIP)

This setting is just optional, as the provider replaces outgoing CLIP which is not valid with the default number of the SIP account.

For sending another CLIP, use the CLIP-No-Screening feature.

Advanced Parameter

Can be empty

Example of VTX gateway settings 2/2.

HomeMenuAdministration

Users & extensions>System>Provisioning>Routes>Gateway groups>SIP gateways>TDM Gateways>Inbound routes>Outbound routes>Call forward>System status>

GigasetProfile

Insecureportinvite - Options "port" and "invite"no - Deactivated (default)no - Deactivated (default)Codes [7]G.722G.729G.711aG.711uGSMH.261H.263H.263+Allowed IP subnet0.0.0.0/0

Advanced parameters

Value

Value

Preview of peer in sip.conf

[gw_1_vtx]
type = peer
host = s1-259634.trk.ipvoip.ch
port = 5060
defaultuser = 413
secret =
outboundproxy = s1-259634.trk.ipvoip.ch
fromdomain =
fromuser =
language = en
insecure = portinvite
sendrpid = no
trustpid = no
nat = yes
directmedia = no
dtmfmode = rfc2833
call-limit = 0
setvar _jic_from_gateway = 1
context = from-gg-vtx
qualify = yes
disallow = all
allow = allow
allow = g722
allow = g729

Inbound routes

In the Gigaset PBX go to "Administration" - "Routes" - "Inbound routes" select the correct Gateway group and press **Show**.

HomeMenuAdministration

Users & extensions>System>Provisioning>Routes>Gateway groups>SIP gateways>TDM Gateways>Inbound routes>Outbound routes>Call forward>System status>

GigasetProfile

Inbound routes

Night answer service

Inbound routes

Gateway groupVTX (vtx)

Advanced optionsOFF

Rule	Number	Target	Profile	
VTX	0041325661425	101	-	
VTX_2	0041325661426	102	-	
			-	

Save

Outbound routes

In the Gigaset PBX "Administration" - "Routes" - "Outbound routes" you don't have to activate the advanced options for the basic installation.

Home

Menu

Administration

Users & extensions

System

Provisioning

Routes

Gateway groups

SIP gateways

TDM Gateways

Inbound routes

Outbound routes

Call forward

System status

Profile

Outbound Routes

Advanced options

Rule

Number

Gateway group

VTX

VTX

VTX

Save