

Interop T440/T640 2.0.0 Deutsche Telefon Trunk

Deutsche Telefon Standard AG Trunk (T440/640 [SW-version 2.0.0](#))

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	Yes
A-number forwarding	Yes
Fax	see chart below

*= Edit the SIP Transports, according to your external IP-address.



- Deutsche Telefon Standard AG Trunk (T440/640 SW-version 2.0.0)
 - Gigaset T440/T640 PRO settings.
 - Gateway Group
 - SIP gateway
 - Inbound routes
 - Outbound routes
 - Fax support

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

Let's assume our SIP trunk contains the following block of numbers:

0891234567[0-9]

And we choose following internal numberblock with 3-digit numbers:

67[0-9]

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.

Edit gateway group

Title: Deutsche Telefon Trunk

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 - Deutsche Telefon Trunk

Outgoing caller IDs

(1) Search/Replace pattern (PCRE) for outbound caller ID signalling. Examples:
Extension only: s/^(\d{3})/\$1/
National 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/^(((0049|0)2631)3370)/
s/^(\d{3})/0049(\d{3})3370(\$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/.

Example of Deutsche Telefon Standard Trunk gateway group settings.

Outbound caller ID

As just the last digit of the trunk number block is changing, you can select just the last digit (in brackets) and put it into the variable \$1. As the provider wants the signalling number in the format +(country-code)(city-code)(number), you have to enter this number into the next line (here: **+49891234567\$1**, where the **\$1** represents the changing part).

The provider will take care about the number representation to the called party. He will present e.g. **08912345678** to the calling party.



Modifications

In case you are using not 3-digit extensions like 123, but 2- or 4-digits, you have to change the Search/replace pattern.

In our example we are using 3-digits, where only the last digit (inside the brackets) is taken into account.

For 2-digits it can look like this:

^\d\d)

+4989123456\$1

Now we're using the complete range from 10 to 99 as extensions, because **\d\d)** is inside the bracket. Therefore we also have to remove the **'7'** from the 2nd line.

When extension **34** is dialing out, the resulting number would be **+498912345634**

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. For Deutsche Telefon Standard Trunk, you can the entries from the outbound caller ID fields e.g. **^\\d\\d(\\d)** and **+49 891234567\$1**

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 pilot-number, according to your line-settings.

^(?:(?:0049|\\+49|49|0)?89)?12345)?(\\d+)\$

This expression cuts all possibly available country- and city-codes and the pilot-number from the incoming number and only the extension is remaining (here: **678**). This will be forwarded to the Inbound Routes.

Inbound caller ID

For incoming calls, we will add an additional **0** in front of the external number of the caller, in order to use the callback-feature of the phone (e.g. **0089987654321**). During the call-setup the additional **0** will be automatically removed.

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 data you received from the provider.

Gigaset

HOME MENU ADMINISTRATION PROFILE

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

Edit SIP gateway : Deutsche Telefon Trunk

Title	Deutsche Telefon Trunk
Name	gw_5_deutschetelefon
Registrar	sip.dtt.de
Proxy [1]	
User [2]	08912345670
Password	*****
Allow outbound calls	<input checked="" type="checkbox"/>
Register	<input checked="" type="checkbox"/>
Language	de - German (de-DE)
Dial command [3]	PJSIP/{prefix}{number.1}@{gateway}
Transport name	default-udp
Source of destination number	INVITE request line
Group [4]	Deutsche Telefon Trunk
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)

(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 PRO will automatically replace {number} by the called number, {number.1} without the first digit and {gateway} with the internal description (e.g. gw_1_amt).

(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](#), [Srv Resource Record \(en\)](#), [CIDR](#), [SIP-DNS-Srv-Records](#), [SIP-DNS-Srv-Records \(en\)](#), [SIP-DNS-Srv-Records \(en\)](#)

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

(7) 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.

Cancel Save

Example of Deutsche Telefon Standard Trunk gateway settings 1/2.

The screenshot shows the Gigaset PBX Administration interface. The sidebar on the left contains navigation links: HOME, MENU, ADMINISTRATION (highlighted), and PROFILE. Below these are sections for Users & extensions, System, Provisioning, Routes, Gateway groups, SIP gateways (highlighted), TDM Gateways, Inbound routes, Outbound routes, Call forwarding, and System status. The main content area is titled 'SIP gateway' and contains several configuration fields:

- DTMF mode: rfc4733 - RTP meta data
- From user: 08912345670
- From Domain: sip.dts.de
- T38 support: OFF
- Update P-Asserted-Identity (CLIP): Use P-Asserted-identity header
- Update remote party ID (CLIP): no - Deactivated (default)
- Trust remote party ID: no - Deactivated (default)
- Codecs [6]: G.722, G.729, G.711a (checked), G.711u, GSM, H.261, H.263, H.263+
- Allowed IP subnet [7]: 0.0.0.0 / 0

Below these fields is an 'Advanced parameters' section with a 'Value' input field and a '+' button. At the bottom is a 'Preview of peer in sip.conf' section showing a sample configuration:

```
[gw_5_deutschetelefon]
type=auth
auth_type=userpass
```

Buttons for 'Cancel' and 'Save' are located at the bottom right of the configuration area.

Example of Deutsche Telefon Standard Trunk gateway settings 2/2.
Registrar

You can use here the DNS-name of the server.

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. **0089987654321**) 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' !!! In case you want to use the prefix-fields from the outbound rules, please add '**{prefix}**' before '**{number:1}**' .

Group

Select here the previously created gateway group.

Update P-Asserted-Identity (CLIP)

To display the correct extension at external parties, you have to use the P-Asserted-Identity (PAI) setting. If this is not selected, only the head-number (e.g. **0891234567-0**) will be presented, no matter which extension is used for dialing.

T.38 support

In order to have the best fax support on the FXS ports, you have to deactivate the T.38 support option. A fallback to G.711 will be used instead.

Advanced Parameter

No advanced parameters necessary.

Underneath you will see then a preview of the sip.conf file.

Inbound routes

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

In addition it is advised to activate the **advanced options** by clicking it to '**ON**' and then pressing **Show**.

The screenshot shows the Gigaset PBX Administration interface. The top navigation bar includes 'HOME', 'MENU', 'ADMINISTRATION' (highlighted), and 'PROFILE'. The left sidebar lists various system settings, with 'Inbound routes' selected. The main content area is titled 'Inbound routes' and features a 'Gateway group' dropdown set to 'Deutsche Telefon Trunk (deutsche-telefon-tru)'. Below this is an 'Advanced options' toggle set to 'ON'. A table lists routing rules with columns for Rule, Active, Date, Weekdays, Time, Profile, Pattern, and Target. Two rules are visible: 'General' and an unnamed rule, both with 'ON' status and active weekdays. The 'General' rule has a pattern of '(*)' and a target of '\$1'. The unnamed rule has a pattern of '' and a target of ''.

Rule	Active	Date	Weekdays	Time	Profile	Pattern	Target
General	ON		M T W T F S S	00:00 to 24:00	-	(*)	\$1
	ON		M T W T F S S	00:00 to 24:00	-		

Example of Deutsche Telefon Standard Trunk Inbound routes.

Rule

Enter here a name for the according rule.

Date / Weekdays / Time

With these settings you can configure a time-controlled routing to different targets.

Pattern / Target

In the pattern you define which part of the incoming number is used to forward the call to the according extension.

In our example we receive already the correct extension from the gateway group. Therefore no further number-manipulation is necessary.

When the PBX finds an according extension it will route the call to it.

But you can add here exceptions from this rule, for example for internal fax users or waiting queues, etc.

Please have in mind the order of these rules, as the system is using **First Match!!!**

Outbound routes

In the Gigaset PBX go to "Administration" - "Routes" - "Outbound routes" activate the Advanced options by clicking it to '**ON**' and then pressing **Show**.

Outbound Routes

Advanced options ☒

Rule	Active	Weekdays	Time	Pattern	Group	Gateway gr...	Add prefix	
General	<input checked="" type="checkbox"/>	M T W T F S S <input checked="" type="checkbox"/> <input checked="" type="checkbox"/> <input checked="" type="checkbox"/> <input checked="" type="checkbox"/> <input checked="" type="checkbox"/> <input checked="" type="checkbox"/> <input checked="" type="checkbox"/>	00:00 to 24:00	*0	[all]	Deutsche		<input type="button" value="-"/> <input type="button" value="0"/> <input type="button" value="."/>
	<input checked="" type="checkbox"/>	M T W T F S S <input checked="" type="checkbox"/> <input checked="" type="checkbox"/> <input checked="" type="checkbox"/> <input checked="" type="checkbox"/> <input checked="" type="checkbox"/> <input checked="" type="checkbox"/> <input checked="" type="checkbox"/>	00:00 to 24:00		[all]	Deutsche		<input type="button" value="+"/>

Example of Deutsche Telefon Standard Trunk Outbound routes.

Rule

Enter here a name for the according rule.

Date / Weekdays / Time

With these settings you can configure a time-controlled routing.

Pattern

In the pattern you define how the outside line is seized. In our example all dialled numbers starting with '0' will use the Gateway group Deutsche Telefon Standard Trunk.

Remark: For calling anonymous to external parties, use the CLIR-setting in the HOME screen of the user!

Fax support

Following table shows you the current state (12.08.2016) of supported fax constellations. These results are without a guarantee. Due to different end-devices, configuration of PBX or other settings the fax transmission might fail. [More info about fax via VoIP networks can be found here.](#)

Receiving	Sending			
	Internal FXS	Internal T38-Fax	External T38-Fax	External machine-Fax
Internal FXS	---	---	OK*	OK*
Internal T38-Fax	---	---	OK**	OK**
External T38-Fax	OK*	OK**	---	---
External machine-Fax	OK*	NOK	---	---

Used devices or services:

Canon Fax-L100 (internal FXS), www.minifax.de (external SW-Fax), Triumph Adler DCC 2725 (external machine-Fax)

* = T38 option in SIP gateway deactivated

** = T38 option in SIP gateway activated