

Interop Hybird 120 Gigaset Edition - Peoplefone

Peoplefone

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

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Hybird 120 Gigaset Edition settings.

In the Hybird 120 Gigaset Edition go to "Assistants" - "PBX" and click on "New": --> add a new SIP Provider

hybird 120 Gigaset

Hybird_120_GE

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English

View

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Go

No.	Name	Connection Type	Ports	Status		
01	Peoplefone	SIP Provider (DDI)	Peoplefone			

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New

PBX - Connections Overview

These assistants guide you through all of the settings required set up and configure an analogue, ISDN or VoIP connection.

All of the configured connections are displayed in the overview. The following are listed: the **Name** of the connection, the **Connection Type** connection type (ISDN, ISDN (P-P), SIP provider, SIP provider (extension) of FXO), the **Ports** used (slot or provider name) and the current **Status**.

The relevant list field is deleted by pressing the button.

You can also edit existing entries with the aid of .

To add a connection, click **New**. Then follow the instructions. The list is called up again once the new connection has been configured, allowing you to set up other connections.

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Provider settings:

Give the Line an unique name, insert the Registration user name and Password.

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Warning: Country code and/or City code not configured!

SIP Provider Settings

Name	Peoplefone
Access Type	Direct Dial-In
Authentication ID	90740928160
Password	
User Name	90740928160
Registrar	app1.peoplefone.ch
Trunk Numbers	
Base Number	091252
Class of Service	
Class of Service	Class of Service CoS Default   Add

Advanced Settings

OK Cancel

PBX - Configuring an SIP provider (extension)

Enter the required data for a "VoIP dial-in with extension" scenario.

Name:
Enter a description for the connection to improve recognition.

Access Type:
DDI is entered here from your previous selection.

Authentication ID:
Enter your provider's authentication ID. A 64 digit alpha-numeric sequence is possible.

Password:
At this point, you can assign a password. A 32 digit alpha-numeric sequence is possible.

User Name:
Enter the user name you received from your VoIP provider. A 64 digit alpha-numeric sequence is possible.

Registrar:
Enter the DNS name or IP address of the SIP server. A 26 digit alpha-numeric sequence is possible.

Base Number:

Just add the telephone numbers for this SIP trunk in the "Advanced Settings"

Advanced Settings

Registrar

Registrar Port

5060

Transport Protocol

☒ UDP
☐ TCP

STUN server

STUN server

Port STUN server

3478

Trunk Numbers

P-P DDI Exception

P-P DDI Exception	Displayed Name	
0570	0912520570	
0571	0912520571	
0572	0912520572	

Add

Further Settings

Generate international phone number

☐ Enabled

Generate national subscriber number

☐ Enabled

OK

Cancel

VoIP settings:

Here you can define how the outgoing lines have to look like (e.g. Clip no Screening)

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SIP Provider

Locations

Codec Profiles

Options

Basic Settings

Description

Peoplefone

Provider Status

☒ Active
 ☐ Inactive

Access Type

☐ Single Number(s)
 ☒ Direct Dial-In

Authentication ID

90740928160

Password

••••••••

User Name

90740928160

Domain

Outgoing Signalling Settings

Outgoing Signalling

Individual CLIP no Screening Number ▾

Signal remote caller number

☒ Enabled

Registrar

Registrar

app1.peoplefone.ch

Registrar Port

5060

Transport Protocol

☒ UDP
 ☐ TCP

STUN

STUN server

Port STUN server

3478

Timer

Registration Timer

60 Seconds

Advanced Settings

OK

Cancel

VoIP Menu Advanced settings:

Please choose here the below settings when you use the CLIP No Screening.

Advanced Settings

Proxy	
Proxy Port	5060
Transport Protocol	☒ UDP ☐ TCP
Further Settings	
From Domain	
Number of allowed simultaneous Calls	No Limitation
Location	Any Location
Codec Profiles	System Default
Dial End Monitoring Time	5 Seconds
Call Hold inside the PBX system	☒ Enabled
Call Forwarding extern (SIP 302)	☐ Enabled
Generate international phone number	☐ Enabled
Generate national subscriber number	☐ Enabled
Deactivate number suppression	☐ Enabled
SIP Header Field for User Name	☐ P-Preferred ☐ P-Asserted ☒ None
SIP Header Field(s) for Caller Address	☐ Display
	☒ User Name
	☐ P-Preferred
	☐ P-Asserted
Substitution of International Prefix with "+"	☐ Enabled
PBX coupling	☐ Enabled
Delete SIP bindings after Restart	☒ Enabled
Upstreaming Device with NAT	☐ Enabled
Early media support	☒ Enabled
Provider without Registration	☐ Enabled