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

+ peoplefone
VoIP Solutions Provider

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

Language English | View Standard | Online Help | Logout

Gigaset

Save configuration

Trunks **Users**

Assistants
- First steps
- Internet Access
- VPN
- **PBX**

System Management
Physical Interfaces
VoIP
Numbering
Terminals
Call Routing
Applications
LAN
Networking
Multicast
WAN
VPN
Firewall
Local Services
Maintenance
External Reporting
Monitoring

View 20 per page « » Go

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

Page: 1, Items: 1 - 1

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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New

Provider settings:

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

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		Displayed Name	
P-P DDI Exception	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)

Save configuration		
Assistants	▼	
System Management	▼	
Physical Interfaces	▼	
VoIP	▲	
Settings		
Numbering	▼	
Terminals	▼	
Call Routing	▼	
Applications	▼	
LAN	▼	
Networking	▼	
Multicast	▼	
WAN	▼	
VPN	▼	
Firewall	▼	
Local Services	▼	
Maintenance	▼	
External Reporting	▼	
Monitoring	▼	

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 Signalisation Settings

Outgoing Signalisation	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