



CLOUD TELECOM AS IT SHOULD BE



Tutoriel de configuration PBX version 66.0
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General ?

Name: OpenIP-Trunk ✓

Type: SIP Registration

Direction: Inbound and outbound

State: Enabled

Global: No

Is secure: No

Inter-Office Trunk ?

Trunk may terminate calls for remote systems: No

Dial plan for outbound calls: Domain Default

Registrations ?

Display name: CompanyName ✓

Account: hEy07t7dnp5c4dPC ✓

Domain: voip.myopenip.fr ✓

Username: hEy07t7dnp5c4dPC ✓

Password: *****

Proxy address: sip:voip.myopenip.fr ✓

Proposed duration (s): 1 hr (Default)

Send re-registrations at this percent of expiry time: 50%

Minimum expiry time: 10 sec

Keep-alive time:

Proxy-Require:

Other headers in registration request:

Send email on status change: Don't send email

DNS lookup for registration refresh: No

Explicitly list addresses for inbound traffic:

Preferred SIP port for trunk:

Routing/Redirection ?

Destination for incoming calls: Send to 10-digit DID

Source for caller-ID: To Header

Default account: 100: AA

Source for caller-ID information: Remote-Party-ID

Fallover behavior: No failover

Request timeout:

Accept redirect: No

Don't accept SIP routing changes in dialog: No

Error message when all lines are busy: 500 Line Unavailable

Redirect destination when all lines are busy: +

Teams: No

Number/Call Identification ?

Prefix:

Trunk ANI:

Regular ANI rule:

Emergency ANI rule:


SIP caller-ID presentation: Custom Headers

Header	Value
Request-URI	Let the system decide (Default)
From	Based on trunk account info
To	Other {request-uri}
P-Asserted-Identity	Don't use header
P-Preferred-Identity	Don't use header
Remote-Party-ID	Based on incoming call
Privacy Indication	RFC3325

Rewrite global numbers: Check domain country code

Country code:

Area code:

P-Charging-Vector	Don't use header
Charging vector	
Diversion header	Don't send
Avoid RFC4122 (UUID)	<input checked="" type="checkbox"/> Yes
Attach PIDF-LO for emergency calls	<input type="checkbox"/> No
Show History-Info	No
Indicate RTPC-XR support	<input type="checkbox"/> No
Encrypt media	Default
Generate unique extension identifier	<input type="checkbox"/> No
Interpret SIP URI always as telephone number	<input type="checkbox"/> No
Caller-ID update on trunk	Don't update
Contact header value	
Customer specific header (Example: X-Vodia-Abc: my header)	
Media/Audio 	
Codec preference	
Active	Inactive
G.711U G.726 GSM 6.10 G.711A	iLBC Opus
<input type="button" value="↑"/> <input type="button" value="↓"/> <input type="button" value="→"/> <input type="button" value="←"/>	
Lock codec during conversation	<input checked="" type="checkbox"/> Yes
Number of codecs in SDP answer	
Generate PCAP traces for calls	<input type="checkbox"/> No
Strict RTP routing	<input checked="" type="checkbox"/> Yes
Port range start	
Port range end	
Trunk requires out of band-DTMF tones	<input type="checkbox"/> No
Requires busy tone detection	<input type="checkbox"/> No
PRACK support	<input checked="" type="checkbox"/> Yes
Ringback on incoming calls	Message 180
Local ringback generation for outbound calls	When there is no SDP in 18x responses

Play file for local ringback when needed	<input checked="" type="checkbox"/> Yes
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