

Inexio SIP Trunk

Hinweise von Inexio

Routes - SIP gateways

Title:	inexio	sip.conf:
Name:	gw_1_inexio	[peer-name]
Registrar / Server:	sip.inexio.net	host
Proxy [N]:		outboundproxy
Username [N]:	496831...@sip.inexio.net	defaultuser / fromuser@fromdomain
Password:		secret
Authuser [N]:		authuser
	<input checked="" type="checkbox"/> permit outbound calls <input checked="" type="checkbox"/> register	... register
Language:	de - German (de-DE)	language = de en
Dial() argument [N]:	SIP/{number:1}@{gateway}	~ Dial(SIP/...)
Extract target number from:	invite - Extract target number from 'invite'	
Group [N]:	inexo	context = from-gg-inexio
Port [N]:	5060 (Default: 5060)	port
Is behind NAT:	<input checked="" type="radio"/> yes <input type="radio"/> no <input type="radio"/> Force rport <input type="radio"/> comedia only (Default: yes)	nat = yes no force_rport comedia
Reroute RTP stream:	no - Don't reroute RTP stream (default)	directmedia = no yes noat update update,soat
Monitor availability:	<input checked="" type="radio"/> yes <input type="radio"/> no (Default: yes)	qualify = yes no
Simultaneous calls:	0 (0 for unlimited, Default: 0)	call-limit
DTMF mode:	rfc2833 - RTP meta data (Default no: rfc2833)	dtmfmode
Insecure:	port,invite - Options 'port' and 'invite'	insecure = no port invite port,invite
Update Remote-Party-ID:	no - Deactivated (default)	sendrpid = no yes pai
Trust Remote-Party-ID:	no - Deactivated (default)	trustrpid = no yes
Codecs:	<input checked="" type="checkbox"/> G.711a <input type="checkbox"/> G.711u <input type="checkbox"/> GSM <input type="checkbox"/> H.261 <input type="checkbox"/> H.263 <input type="checkbox"/> H.263+ (Default: G.711a)	allow
Permit IP subnet [N]:	0 / 0 / 0 / 0 (Default: 0.0.0.0)	permit

Routes - SIP gateways

Title:	inexio 2	sip.conf:
Name:	gw_1_inexio2	[peer-name]
Registrar / Server:	sip.inexio.net	host
Proxy [N]:		outboundproxy
Username [N]:	496831...@sip.inexio.net	defaultuser / fromuser@fromdomain
Password:		secret
Authuser [N]:		authuser
	<input checked="" type="checkbox"/> permit outbound calls <input checked="" type="checkbox"/> register	... register
Language:	de - German (de-DE)	language = de en
Dial() argument [N]:	SIP/{number:1}@{gateway}	~ Dial(SIP/...)
Extract target number from:	invite - Extract target number from 'invite'	
Group [N]:	inexo	context = from-gg-inexio
Port [N]:	5060 (Default: 5060)	port
Is behind NAT:	<input checked="" type="radio"/> yes <input type="radio"/> no <input type="radio"/> Force rport <input type="radio"/> comedia only (Default: yes)	nat = yes no force_rport comedia
Reroute RTP stream:	no - Don't reroute RTP stream (default)	directmedia = no yes noat update,soat
Monitor availability:	<input checked="" type="radio"/> yes <input type="radio"/> no (Default: yes)	qualify = yes no
Simultaneous calls:	0 (0 for unlimited, Default: 0)	call-limit
DTMF mode:	rfc2833 - RTP meta data (Default no: rfc2833)	dtmfmode
Insecure:	port,invite - Options 'port' and 'invite'	insecure = no port invite port,invite
Update Remote-Party-ID:	no - Deactivated (default)	sendrpid = no yes pai
Trust Remote-Party-ID:	no - Deactivated (default)	trustrpid = no yes
Codecs:	<input checked="" type="checkbox"/> G.711a <input type="checkbox"/> G.711u <input type="checkbox"/> GSM <input type="checkbox"/> H.261 <input type="checkbox"/> H.263 <input type="checkbox"/> H.263+ (Default: G.711a)	allow
Permit IP subnet [N]:	0 / 0 / 0 / 0 (Default: 0.0.0.0)	permit

Bei Inexio wird die Ausgehene Rufnummer per P-Preferred-Identity Signalisiert. Damit die UCware P-Preferred-Identity verwendet muss folgende Datei in /etc/ucware/asterisk/quirks.d hinterlegt sein:

p-preferred-identity.lua

```
local Channel = require 'core.channel'  
local Log = require 'core.logger'  
  
local cid = Channel["CALLERID(num)"]  
app.sipaddheader("P-Preferred-Identity: <sip:..cid.."@inexio.de>")
```

From:

<https://wiki.ucware.com/> - **UCware-Dokumentation**

Permanent link:

<https://wiki.ucware.com/gateway/sip/inexio>

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