

Konfiguration

Wie richte ich mein snom VoIP Telefon ein?

Zur Einrichtung ihres snom VoIP Telefons brauchen Sie die VoIP Zugangsdaten. Sollten Sie diese noch nicht gefunden haben, hilft Ihnen dieser FAQ Eintrag:

[Wo finde ich meine VoIP Zugangsdaten?](#)

Bitte tragen Sie die VoIP Zugangsdaten folgendermaßen ein und nehmen die folgenden Konfigurationen vor:

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SIP Identity Settings:

Music on hold server:	<input type="text"/>	?
Send hold as inactive:	<input type="radio"/> on <input checked="" type="radio"/> off	?
Alert Info URL:	<input type="text"/>	?
User picture URL:	<input type="text"/>	?
Dial-Plan String:	<input type="text"/>	?
Count all groups in Dial-Plan:	<input type="radio"/> on <input checked="" type="radio"/> off	?
ENUM Support:	<input type="radio"/> on <input checked="" type="radio"/> off	?
Countrycode:	<input type="text"/>	?
Areacode:	<input type="text"/>	?
Proxy Require:	<input type="text"/>	?
Additional supported headers:	<input type="text"/>	?
Q-Value:	1.0 <input type="text"/>	?
Proposed Expiry:	3600	?
Auto Answer:	<input type="radio"/> on <input checked="" type="radio"/> off	?
Long SIP-Contact (RFC3840):	<input checked="" type="radio"/> on <input type="radio"/> off	?
Support broken Registrar:	<input checked="" type="radio"/> on <input type="radio"/> off	?
Shared Line:	<input type="radio"/> on <input checked="" type="radio"/> off	?
Publish Presence on bootup:	<input checked="" type="radio"/> on <input type="radio"/> off	?
DTMF via SIP INFO:	off <input type="text"/>	?
Send display name on INVITE:	<input type="radio"/> on <input checked="" type="radio"/> off	?
Extension Monitoring Call Pickup List URI:	<input type="text"/>	?
Contact List:	<input type="radio"/> on <input checked="" type="radio"/> off	?
Publish Presence:	<input type="radio"/> on <input checked="" type="radio"/> off	?
Contact List URI:	<input type="text"/>	?
Force sendrecv on INVITE with no SDP:	<input type="radio"/> on <input checked="" type="radio"/> off	?
Server Type Support:	Default <input type="text"/>	?
Remove all bindings on unregister:	<input type="radio"/> on <input checked="" type="radio"/> off	?
Subscription Expiry (s):	3600	?
Failed Subscription Retry Time (s):	600	?
Enable hook flash:	<input type="radio"/> on <input checked="" type="radio"/> off	?
Identity can receive calls:	<input checked="" type="radio"/> on <input type="radio"/> off	?
Allow incoming extension monitoring:	<input checked="" type="radio"/> on <input type="radio"/> off	?
Extension monitoring group ID:	<input type="text"/>	?
Device Feature Key Synchronisation	<input type="radio"/> on <input checked="" type="radio"/> off	?

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NAT Identity Settings:

Offer ICE:

on off ?

STUN server (IP-addr:port):

5.9.87.18:3479 ?

STUN interval (seconds):

3600 ?

Keepalive interval (seconds):

30 ?

Apply

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RTP Identity Settings:

Codec:

pcmu,pcma,g726-32,aal: ?

Packet Size:

20 ms ?

Filtered codec list:

pcmu, pcma, g726-32, aal2-g726-32, g729, telephone-event

Full SDP Answer:

on off ?

Symmetrical RTP:

on off ?

RTP Encryption:

on off ?

G.726 Byte Order:

RFC3551 AAL2 ?

SRTP Auth-tag:

AES-32 AES-80 ?

RTP/SAVP:

off ?

Media Transport Offer:

UDP ?

Media Transport Offer Setup:

active ?

Multicast relay address:

?

Apply

Eindeutige ID: #1374

Verfasser: Service Team

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