

Network

Telephony

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DTMF over VoIP connections

Send settings:

 Automatic Audio RFC 2833 SIP Info

When using G.722-Codex (wide-band connection) DTMF Signals cannot be transmitted via audio.

Call Transfer

Use the R key to initiate call transfer with the SIP Refer method:

 Yes No

Transfer Call by On-Hook:

 Yes No

You can define the choice of target address in the SIP protocol.

Find target addr. automatically:

 Yes No

Derive target address:

 from the SIP URL from the SIP contact header

Hold on transfer target:

 For attended transfer
 For unattended transfer

Hook Flash (R-key)

R-key settings are disabled because the R key is used for call transfer.

Listen ports for VoIP connections

Use random ports:

 Yes No

SIP port:

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RTP port:

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