

DTMF support

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Overview

DTMF - Dual-Tone Multi-Frequency signaling

This signaling method is used for telecommunication over analogue telephone lines in the voice-frequency band between telephone handsets and other communication devices and the switching center.

Methods of sending DTMF

There are 3 methods of sending DTMF in the SIP environment:

1. SIP INFO packets.
2. As specially marked events in the RTP stream - see: [RFC 2833](#).
3. Inband as normal audio tones in the RTP stream with no special coding or markers.

Flashphoner supports DTMF with 1 and 2.

DTMF settings

Using SIP INFO packets

To send DTMF using SIP INFO packets, the following parameter in [flashphoner.properties](#) file should be set

```
dtmf=INFO
```

PBX configuration

To receive DTMF using SIP INFO packets, PBX should be configured as follows (Asterisk for example)

```
[general]
bindport=5060
bindaddr=0.0.0.0
context=default
dtmfmode=info
allow=all

[2000]
type=friend
secret=2000
host=dynamic
canreinvite=no
dtmfmode=info
```

Note that Asterisk uses `inband` method by default which does not supported by WCS.

DTMF over RFC2833

To send DTMF as specially marked events according to RFC 2833, the following parameter in [flashphoner.properties](#) file should be set

```
dtmf=RFC2833
```

Also, telephone-event codec should be enabled, for example

```
codecs                =opus,alaw,ulaw,g729,speex16,g722,mpg4-generic,telephone-event,h264,vp8,flv,mpv
codecs_exclude_sip    =mpeg4-generic,flv,mpv,alaw,ulaw,g729,speex16,g722,vp8
codecs_exclude_streaming =telephone-event
codecs_exclude_sip_rtmp =opus,g729,g722,mpg4-generic,vp8,mpv
```

In the case above, H264 for video, opus for audio and telephone-event for DTMF are enabled.

PBX configuration

To receive DTMF according to RFC 2833, PBX should be configured as follows (Asterisk for example)

```
[general]
bindport=5060
bindaddr=0.0.0.0
context=default
dtmfmode=rfc2833
allow=all

[2000]
type=friend
secret=2000
host=dynamic
canreinvite=no
dtmfmode=rfc2833
```

Note that Asterisk uses inband method by default which does not supported by WCS.