

# WebRTC-SIP gateway functions

- [DTMF support](#)
- [SIP calls in a WebRTC-compatible browser](#)
- [SIP calls using Android SDK](#)
- [SIP calls using iOS SDK](#)
- [SIP complex functions](#)
  - [A call between two browsers made via the SIP server](#)
  - [A call to a SIP conference](#)
  - [A call to a mobile phone via the SIP server](#)
  - [Known issues](#)