



Create WebRTC Voice Applications with Microsoft .NET (C#/VB.NET)

Voice Elements Platform, our powerful SIP platform and programmable .NET API now supports WebRTC. Built right into our included SIP stack, is a bridge that connects WebRTC to SIP and will enable voice applications built with Voice Elements to make and receive calls using WebRTC.

Voice Elements Platform supports an impressive array of voice processing via the WebRTC connection.

Voice Elements Platform with WebRTC Features:

- Microsoft .NET (C#/VB.NET)
- WebRTC Codecs: G.711, OPUS, ISAC over SRTP
- RTP Presence for WebRTC
- Conferencing
- Speech Recognition and Text-to-Speech (TTS)
- Beep Detection
- Highly Scalable
- WEBRTC and SIP Call Control
- SIP Codecs: G.711, G.729 over SRTP
- Voice (Play, Record, Get Digits)
- Faxing
- Highly Accurate Call Analysis
- Runs on Windows

In addition, the data channel has been enabled which provides a rich interface between a browser- based application and Voice Elements Platform. The applications are limitless!

What can you build with Voice Elements Platform?

You can build any voice application with Voice Elements Platform via SIP or WebRTC. Voice Applications from Toll Bypass, Call Center apps, Virtual Employee apps to sophisticated IVRs and Voice Dialers, Locator apps and much more. In addition, Voice Elements Platform support speech recognition and TTS and award-winning Call Progress Analysis.

For pricing and a free demonstration contact Inventive Labs sales directly at sales@inventivelabs.com.
www.VoiceElements.com or call us at 866-923-5290