

GENESYS

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Web Real-Time Communications Deployment Guide

Codec Support

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The Genesys WebRTC Gateway has the following codec support:

Codec	Codec Type	Supported Interfaces/ Protocols
G.711	Audio	SIP, WebRTC
G.722	Audio	SIP, WebRTC
G.729	Audio	SIP
telephone-event	Audio/DTMF	SIP, WebRTC
VP8	Video	SIP, WebRTC
H264	Video	SIP