

Open telephony bridges

A SIP trunk joins two dial plans. It does not merge them.

SIPconnect is the SIP Forum recommendation for provider peering.

Asterisk and FreeSWITCH are separate PBXs with their own dialplans.

WebRTC is a browser API plus the protocol suite in RFC 8825. SIP over WebSocket is RFC 7118.

A CPaaS vendor matrix was not opened. Confirm a vendor on that vendor's page.

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Sources and go deeper

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