LiveSwitch

MASSIVELY FLEXIBLE, HIGHLY SCALABLE
VIDEO CONFERENCING SDK & SERVER STACK









The Problem

In today's digital marketplace, consumers have come to expect high-quality live audio and video experiences in customized, user-friendly applications. Yet, the live video conferencing solutions on the market today often lack the flexibility needed to create these rich experiences, interop with legacy applications, or accommodate less-than ideal network conditions and device performance.



The Solution

LiveSwitch is a massively flexible, highly scalable video conferencing SDK and server stack capable of dynamically and simultaneously connecting two or more WebRTC remote candidates into the cost-optimized conferencing mode best suited for your business. Specifically designed to be the most flexible enterprise-grade professionally supported SDK on the market today, LiveSwitch provides everything your team needs to build a custom application that fully integrates voice, video, chat, IoT technology, telephony, broadcasting, and more into your existing applications.



Benefits

- LiveSwitch's flexible architecture reduces costs by providing an API for dynamically selecting the optimal connection type for each participant.
- LiveSwitch provides the core technology needed for many kinds of live video use cases, including kiosks, augmented/virtual reality, surveillance and other special IoT devices.
- LiveSwitch supports all browsers, mobile and native platforms.
- LiveSwitch is a self-hosted software based on-premise solution that allows you to choose when and where to deploy your technology.
- LiveSwitch handles all the complexities of signaling, codec selection, transcoding, and network bandwidth adaptation, letting your team focus on the important stuff.

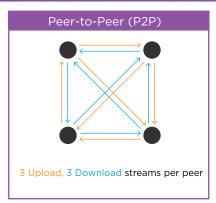
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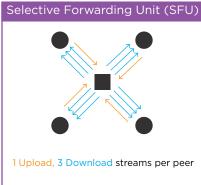
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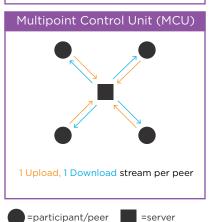
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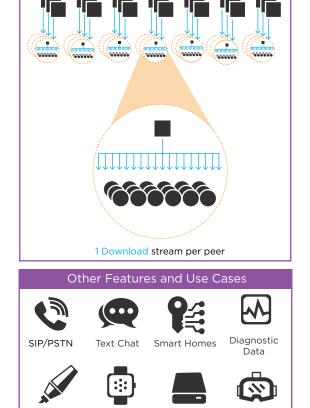












Sub-Second Latency Broadcasting

< How It Works

These diagrams are an example of how *LiveSwitch* can be flexibly used in multiple modes

Hybrid Topology

LiveSwitch can seamlessly switch between P2P, SFU and MCU connections as needed by the number of participants or device limitations.

Broadcasting

LiveSwitch allows WebRTC enabled unidirectional video broadcasts that transmit one-to-many audio/video feeds through a cluster of media servers with sub-second latency.

Telephony Integration

LiveSwitch fully integrates with SIP, VoIP and PSTN to allow participants to connect via telephone to an existing WebRTC-based video conference.

The WebRTC Solution that Plays Nice with Everyone

and MORE!

Wearables











Whiteboards





Recording Virtual Reality







