

Videoconference system for E-learning platform based on WebRTC protocol



About presentation

-  **WebRTC – Definition and usage**
-  **System architecture**
-  **Centralization of media streams**



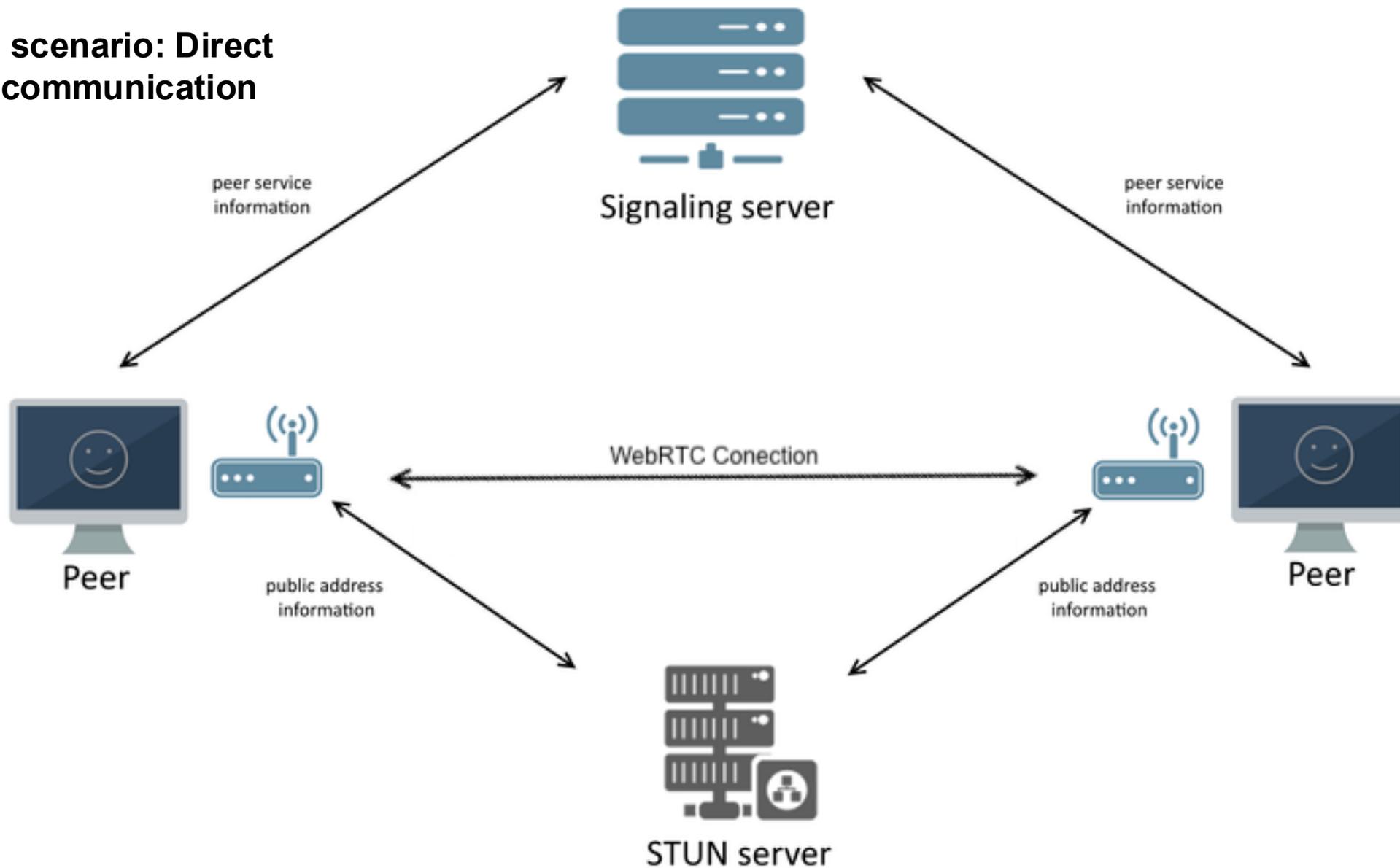
WebRTC: Definition



WebRTC (Web Real Time Communication) is an open source technology and set of protocols that enables real-time communication – including audio, video and arbitrary data directly between browsers and devices using peer-to-peer (P2P) connections

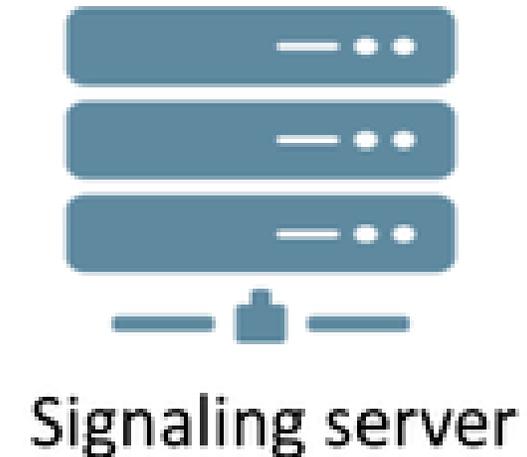


**Ideal WebRTC scenario: Direct
Peer-to-Peer communication**

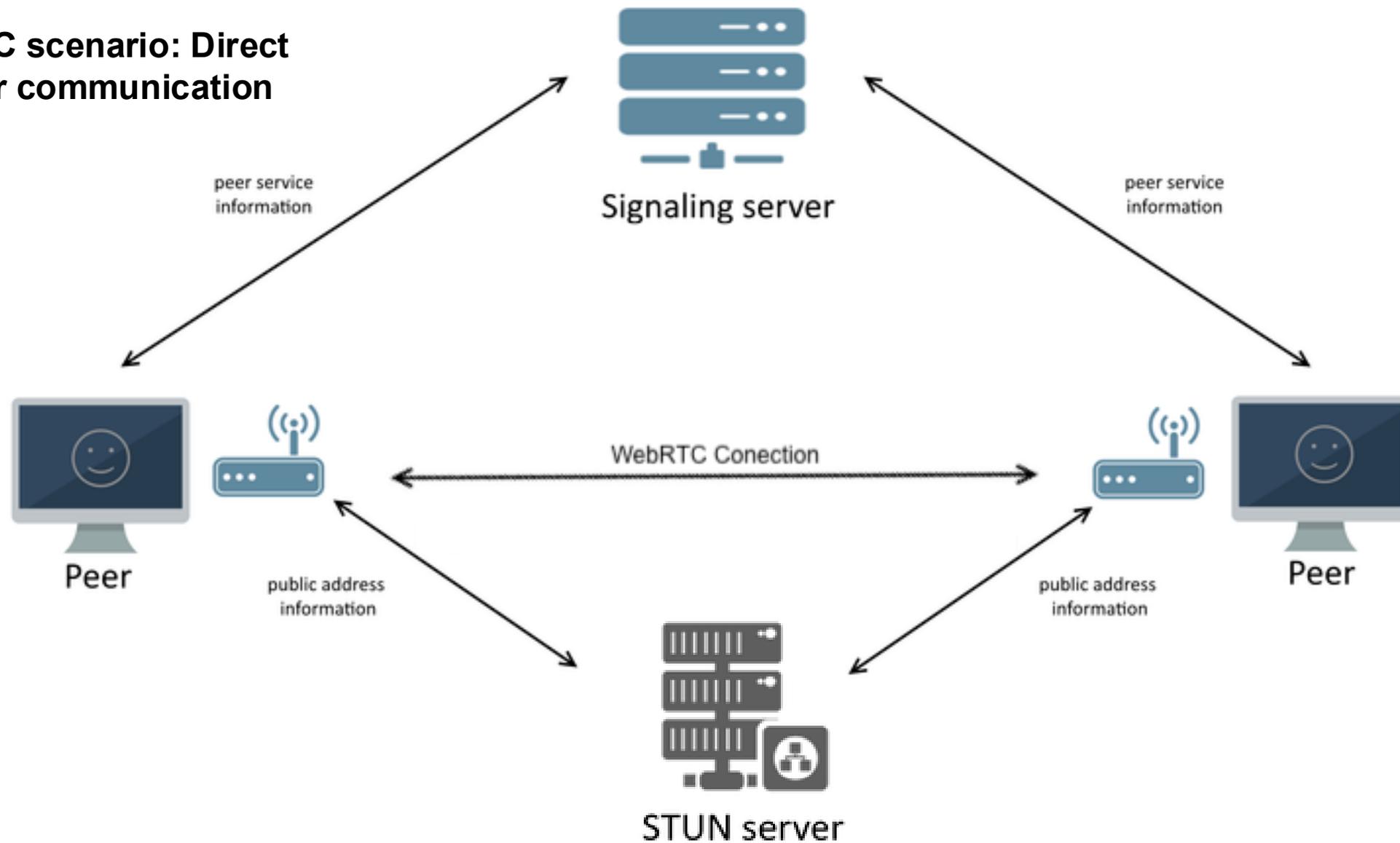


Signaling Server

-  It's purpose is to exchange connection-related information between 2 peers
-  It is commonly built on top of WebSocket technology to enable low-latency communication
-  The exchanged connection data includes: the SDP offer/answer and ICE candidates

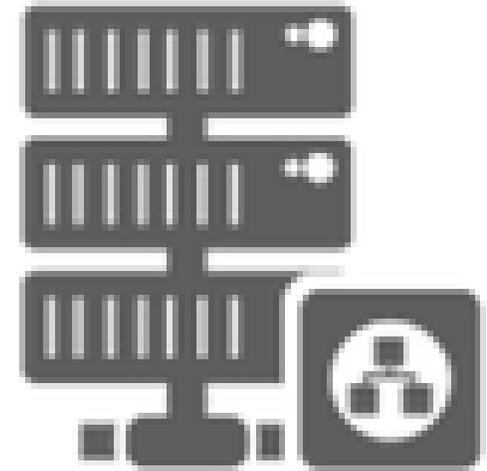


Ideal WebRTC scenario: Direct Peer-to-Peer communication



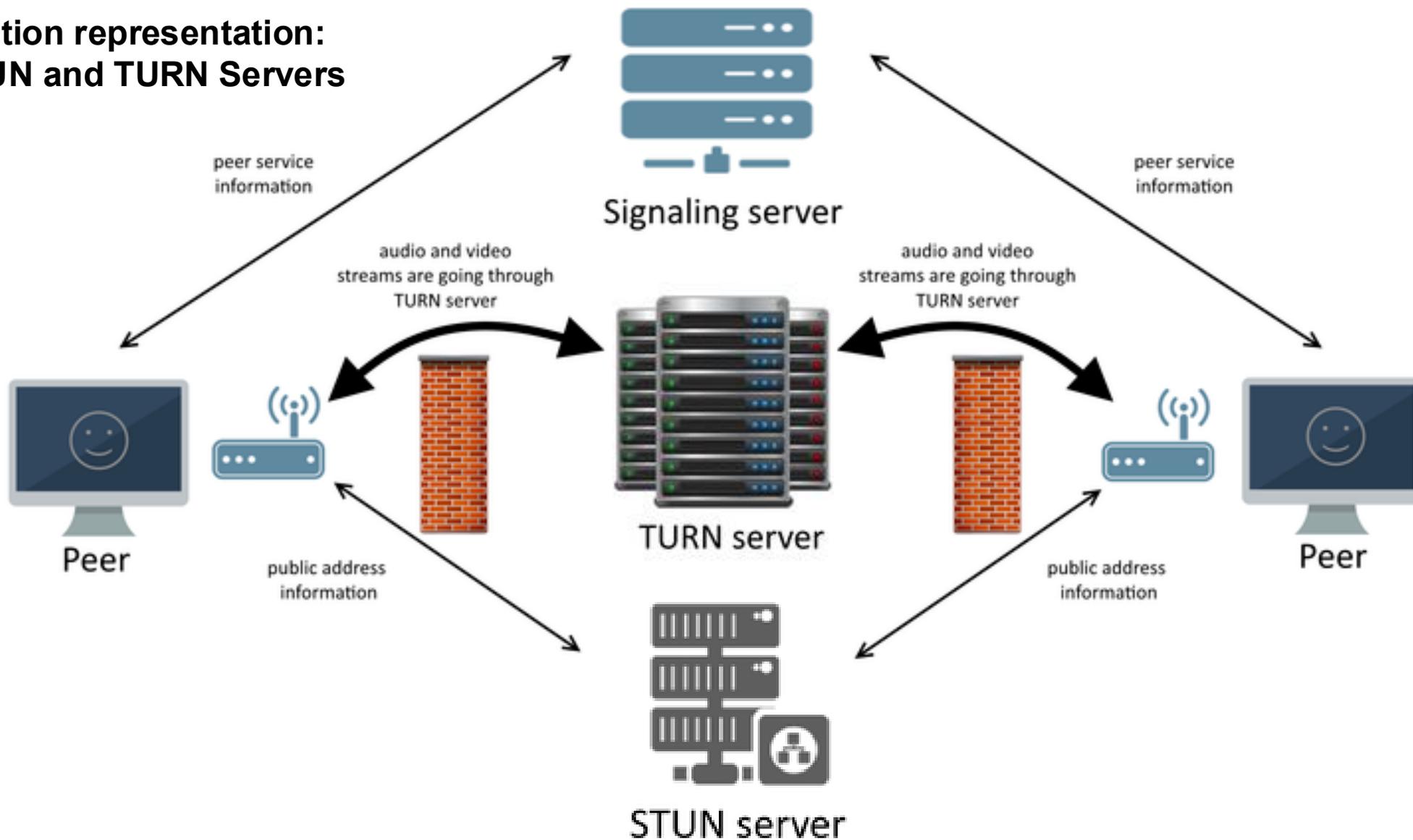
STUN: Session Traversal Utilities for Nat

-  It's purpose is to determine the public IP addresses of the peers involved in the communication
-  The server is required only during the initial connection setup between peers
-  Beside public IP, it also delivers information about the type of NAT configuration the device is behind



STUN server

WebRTC connection representation: Scenario with STUN and TURN Servers



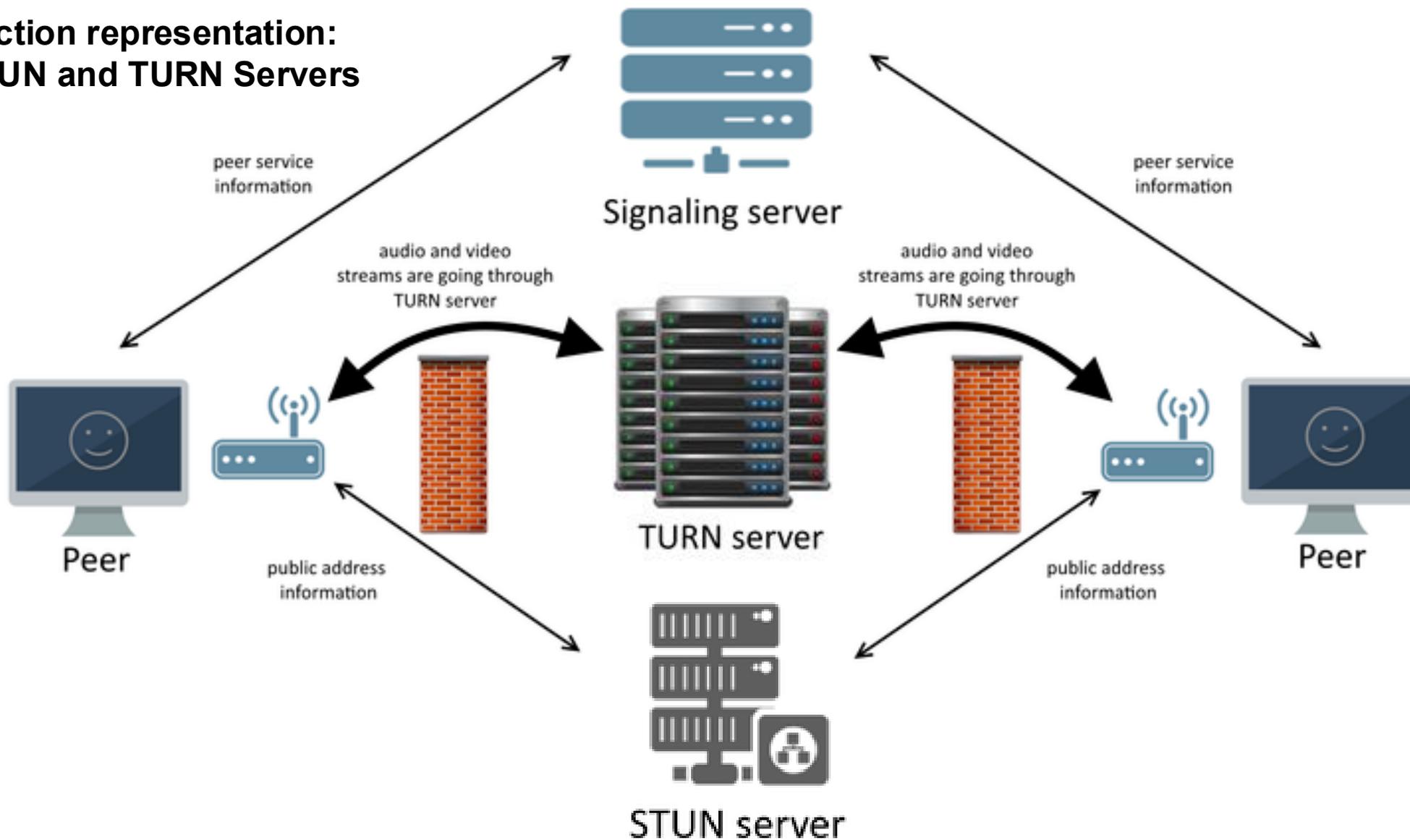
TURN: Traversal Using Relays around NAT

-  The TURN server's role is to facilitate and intermediate data relaying between participants
-  TURN is used as a fallback mechanism when direct P2P connection is impossible
-  Built to handle various network conditions, it enables adaptive media streaming over both TCP/IP and UDP, ensuring flexibility and reliability

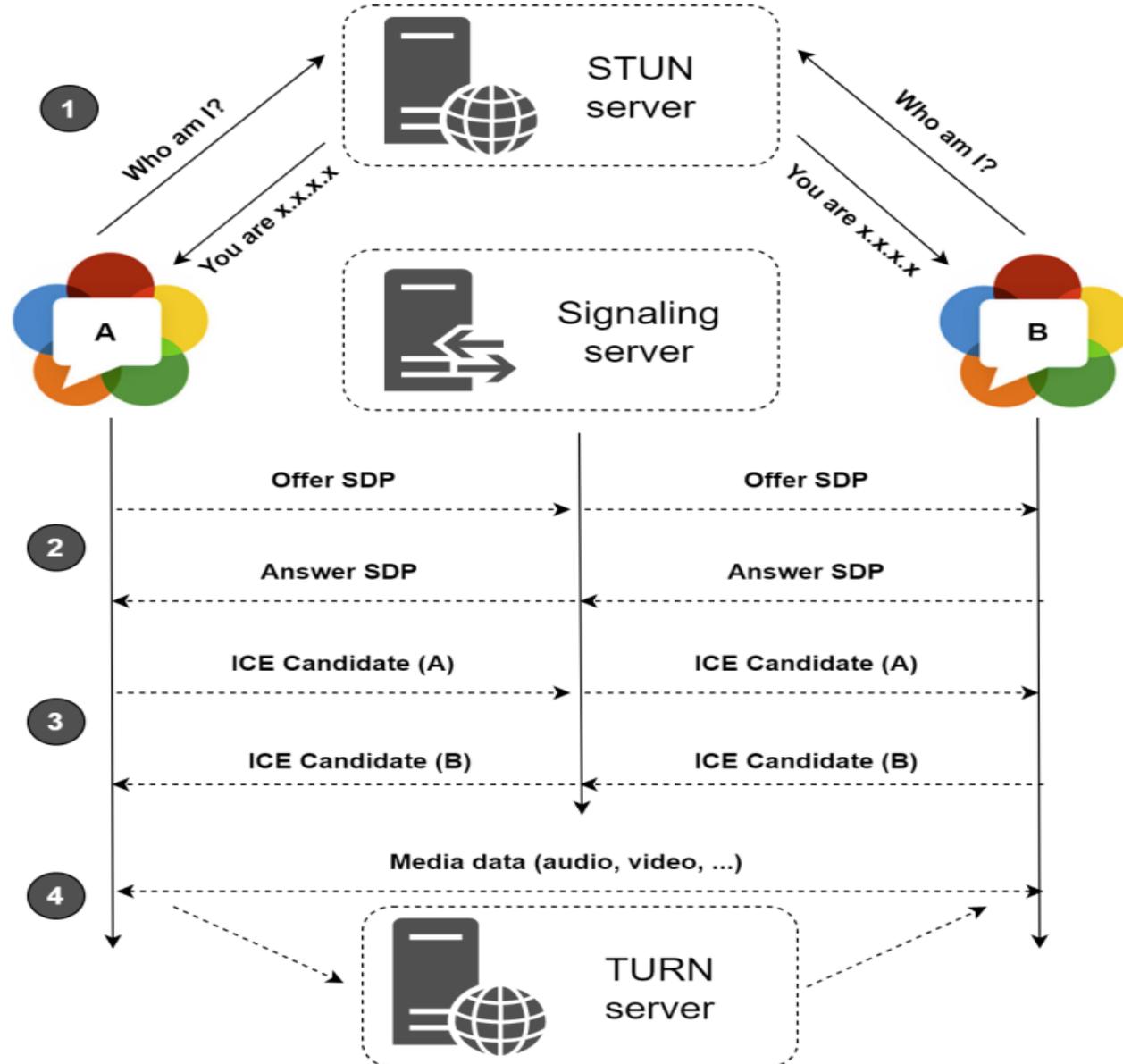


TURN server

WebRTC connection representation: Scenario with STUN and TURN Servers



Connection setup Diagram: Step-by-step WebRTC Flow



Centralisation of Media streams

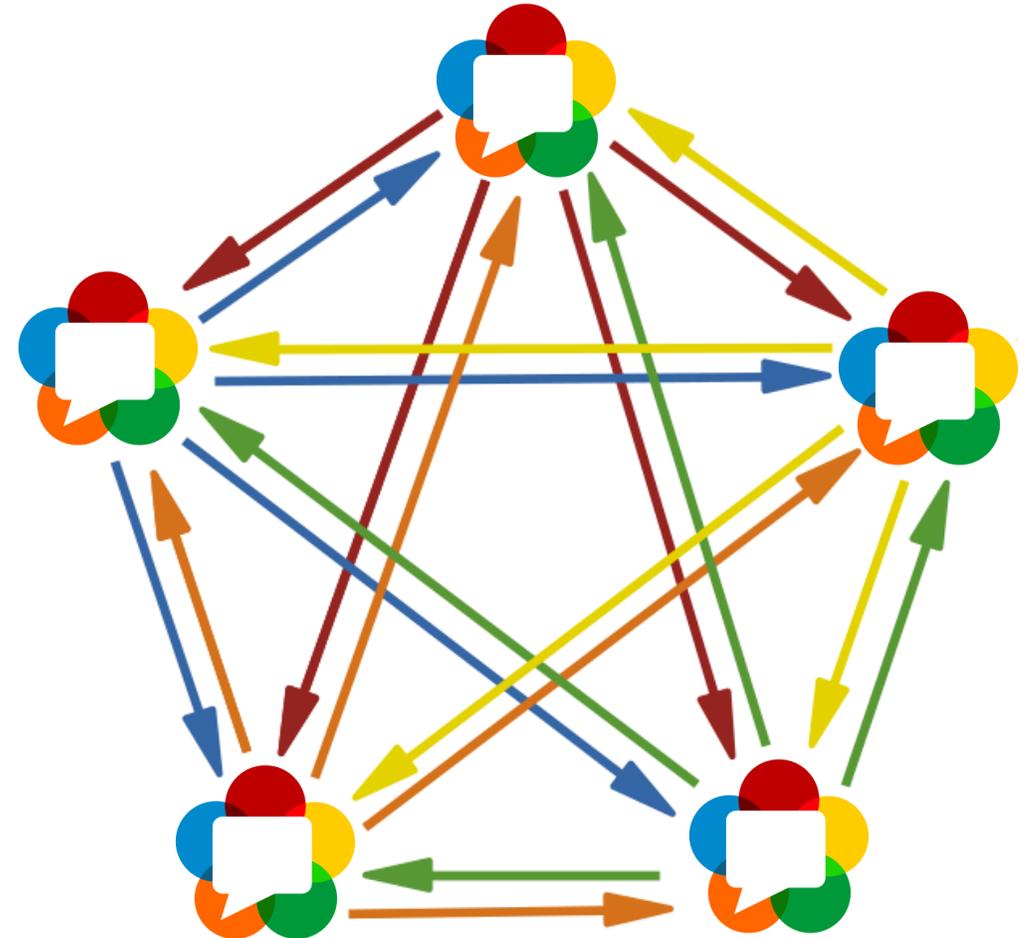
 The implementation of E-Learning video conferencing system can rely on three main methods of media centralization and distribution



Centralisation of Media streams

Mesh Topology

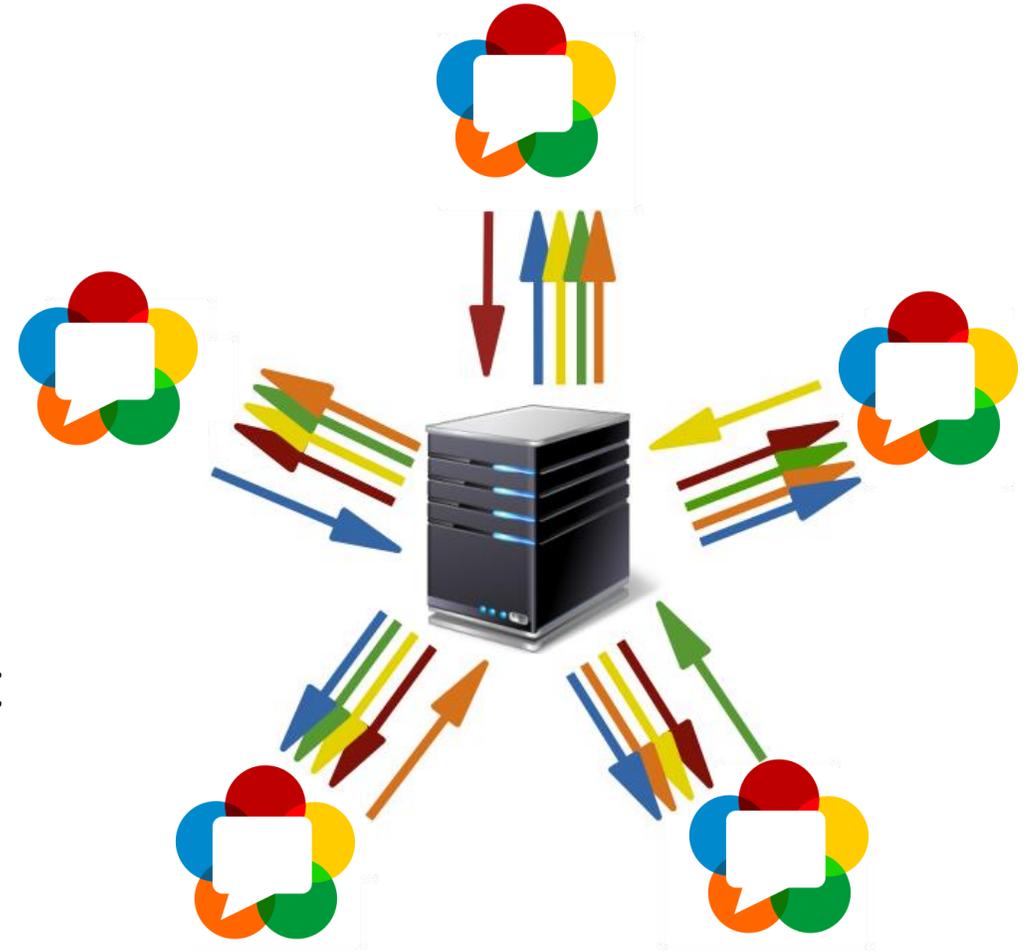
This model assumes a decentralized system where streams are transmitted directly between participants. It is used for conferences with a small number of users (approximately 3-5)



Centralisation of Media streams

SFU – Selective Forwarding Unit

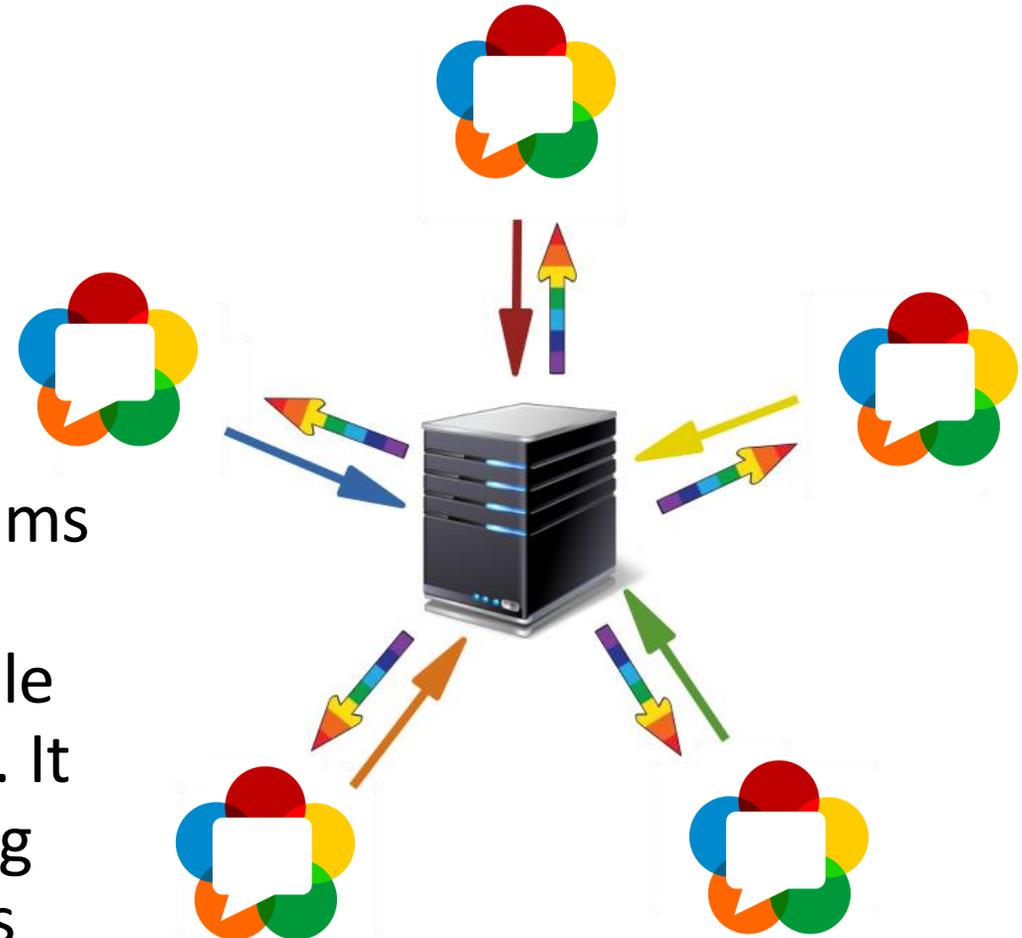
This model centralizes the streams, but not in the most efficient way. It allows connecting multiple participants, but yet it is not well suited for E-Learning platform



Centralisation of Media streams

MCU – Multipoint Central Unit

This model not only centralizes the streams but also optimizes their distribution by compositing multiple streams into a single one and retransmitting it to participants. It is optimal solution for video conferencing with a large number of users, which suits our needs perfectly



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