



SWITCH

The Swiss Education & Research Network

Basic Architecture of H.323

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- **Background to H.323**
- **Components of H.323**
- **H.323 Protocols Overview**
- **H.323 Call Establishment**

- **ITU-T¹ published Version 1 of Recommendation H.323 in 1996:**

“Visual Telephone Systems and Equipment for LANs which provide a non-guaranteed Quality of Service”

Æ this standard was not designed for the *Internet*
(Bandwidth/QoS of Internet links did not allow for transmitting video streams in real-time)

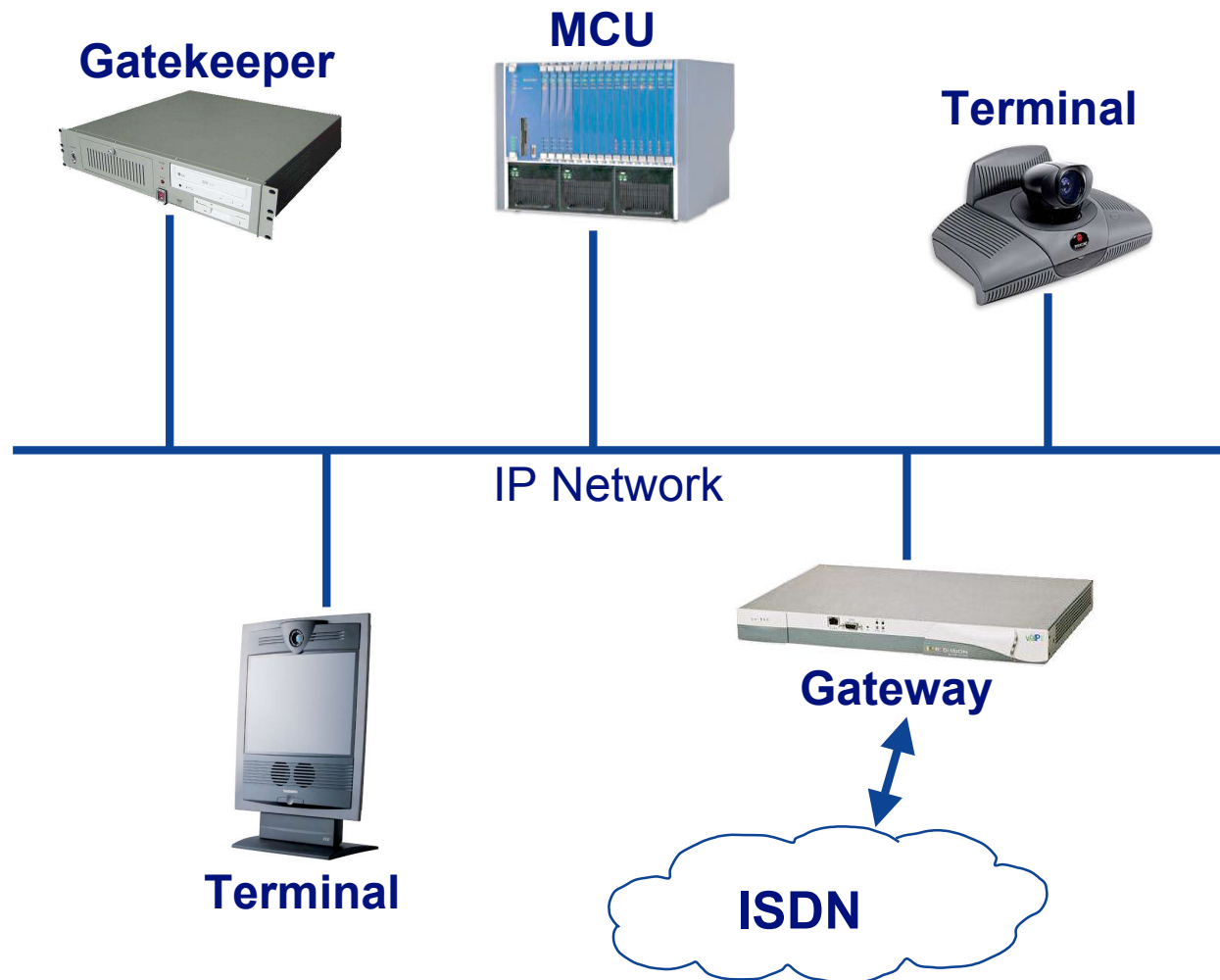
Æ Internet scalability issues were neglected
(only local calls, small number of users)

¹ TELECOMMUNICATION STANDARDIZATION SECTOR OF ITU

- **H.323 borrowed much of the multimedia conferencing aspects from other H.32x-series² Recommendations, such as H.320**
 - Æ good interoperation with ISDN
 - Æ bad interoperation with Internet Standards (security, addressing)
 - **The name was changed in Version 2 (1998) to “*Packet-Based Multimedia Communications Systems*”**
 - **In November 2000 the most recent Version 4 of H.323 was published**
- ² H.320: N-ISDN Videoconferencing
H.321: B-ISDN (ATM) Videoconferencing
H.322: Guaranteed QoS LAN (e.g. Iso-Ethernet) Videoconferencing
H.324: GSTN Videoconferencing (low bit-rate multimedia communication)
Æ over modem links

- H.323 defines the *interworking of*
 - *call signaling,*
 - *call control,*
 - *and media stream protocols,*in order to build a packet-based multimedia communications system
- H.323 further describes the *network components* that are used to build up such a communications system
- H.323 can be seen as an “umbrella standard” which aggregates standards for multimedia conferencing over packet-based networks

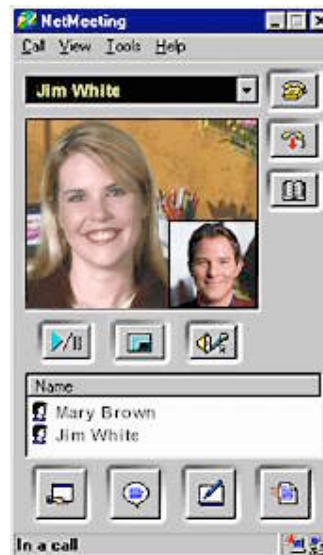
H.323 Components



- Terminal
 - Video/audio/data client
- MCU
 - Conference control
 - Content mixing
- Gateway
 - Protocol translation
- Gatekeeper
 - Address resolution
 - Admission control
- Terminals, MCUs, and Gateways are called *H.323 Endpoints*
- An endpoint is “callable”

H.323 Terminals

- An endpoint on the network which provides for *real-time, two-way communications with another H.323 terminal, GW, or MCU*
- This communication consists of speech only, speech and data, speech and video, or speech, data and video

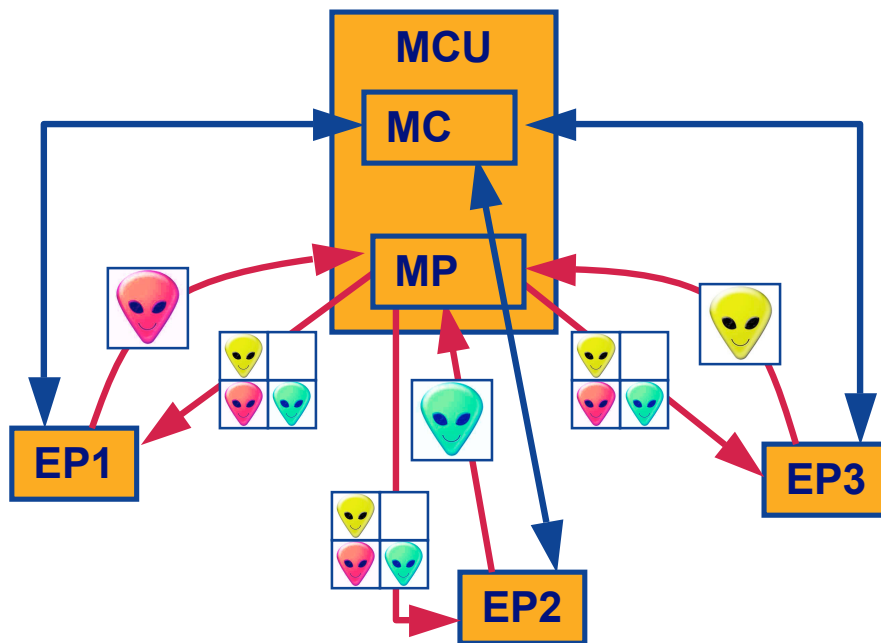


Multipoint Control Unit (MCU)

- An MCU consists of a mandatory *Multipoint Controller* (MC) and an optional *Multipoint Processor* (MP)

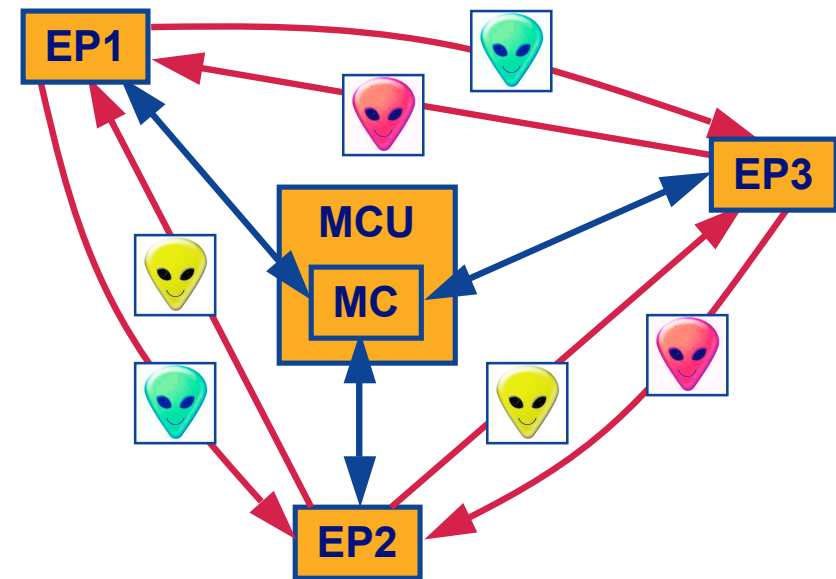
MC: call signaling, conference control
MP: switching/mixing of media streams

Centralized multipoint conference



unicast media streams

Decentralized multipoint conference



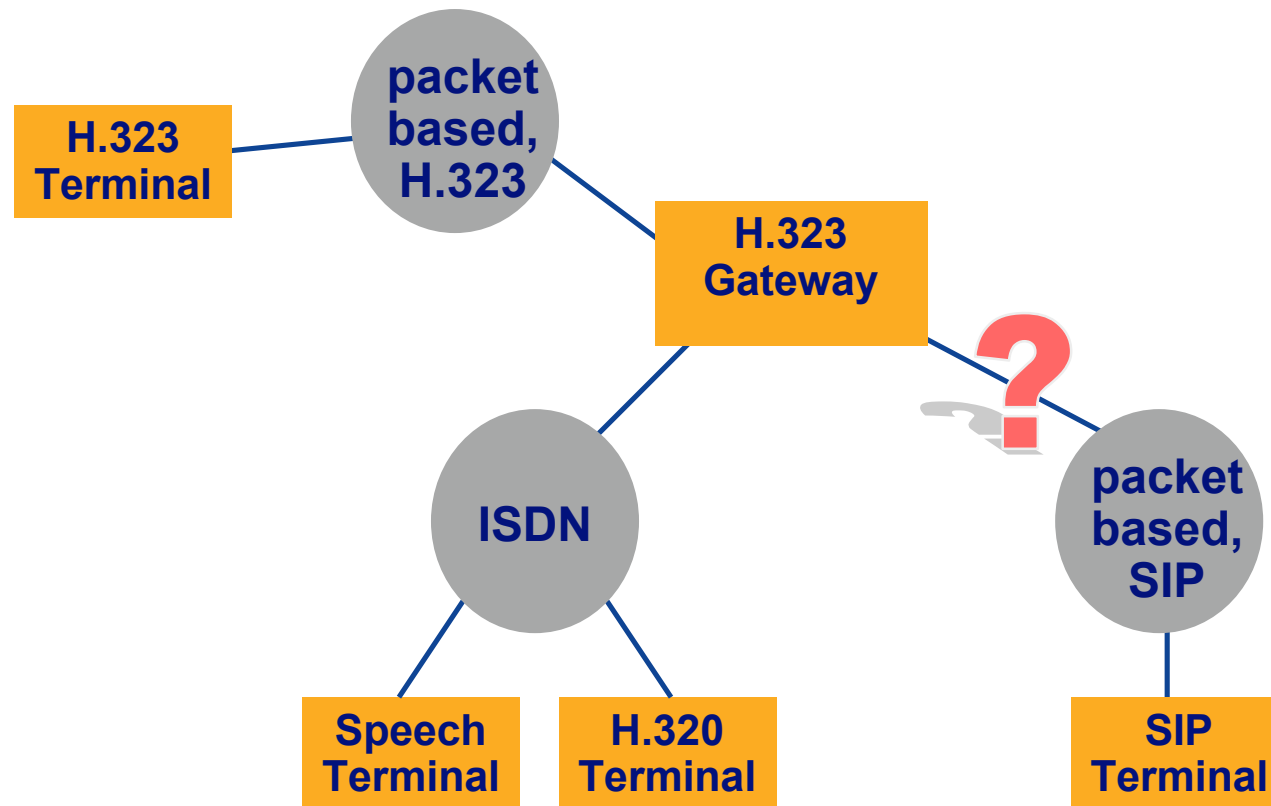
multicast media streams

- **Some MPs can do *real-time transcoding* of the received audio/video streams**
 - Æ every participating terminal can choose its preferred audio/video codec
 - Æ transcoding of video streams is computationally expensive, so this is normally done in dedicated DSPs



MCU with real-time transcoding (Accord MGC-100)

- **H.323 Gateway = H.323 endpoint which provides for real-time, two-way communications between *terminals belonging to networks with different protocol stacks***



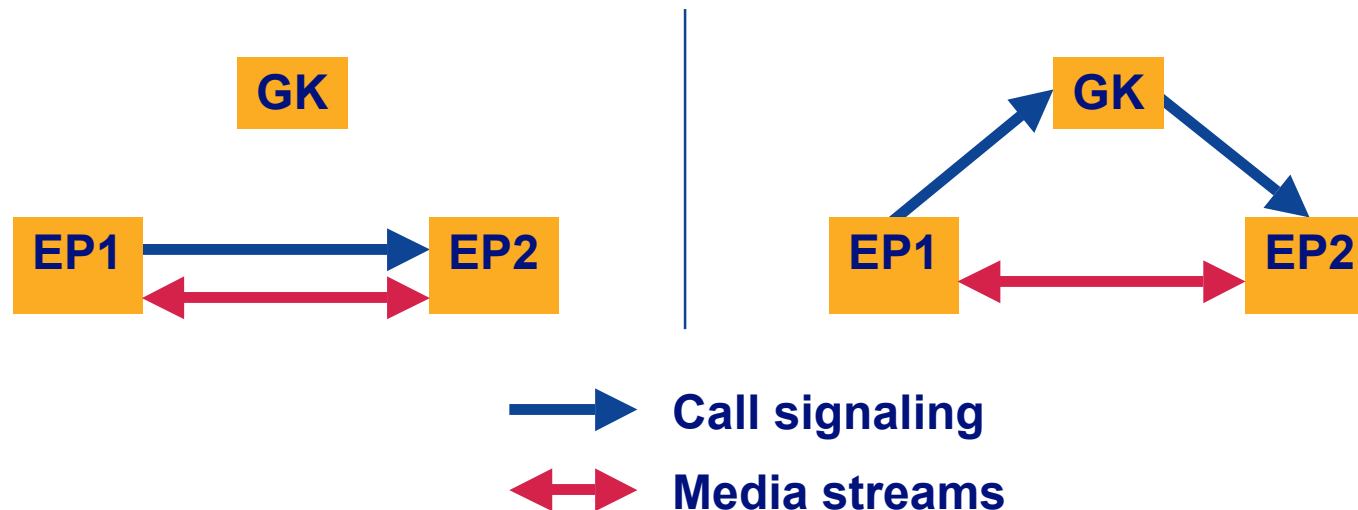
- **A GK is an H.323 entity on the network that provides *address translation and controls access to the network resources* for H.323 terminals, GWs and MCUs**
- **Endpoints do register themselves at a GK**
- **All H.323 endpoints registered to a single GK build an *H.323 zone***
 - **H.323 zones are independent of physical network topology**
 - **Each zone has only one GK (exception: Alternate GKs)**

- **Address translation:**
 - No “normal” DNS address resolution is possible
 - Endpoints do register with their H.323 aliases¹ and call signaling IP address
 - A GK translates H.323 aliases into call signaling IP addresses (especially useful for endpoints with dynamic IP addresses)
 - Multiple GKs can communicate to build a multi-zone address translation service (Æ e.g. Global Dialing Scheme, GDS)

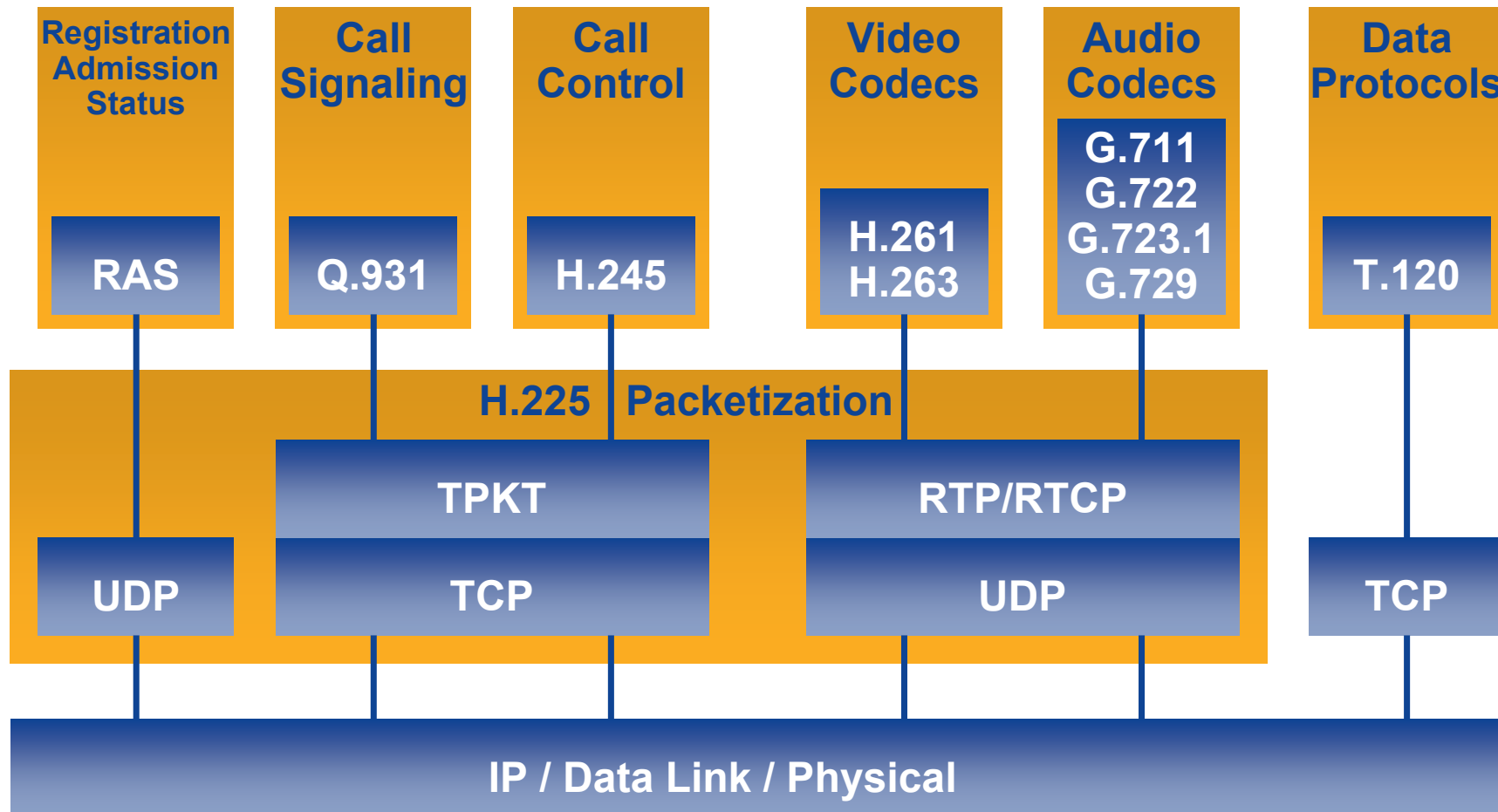
¹ H.323 alias = email-ID (e.g. schlatter@switch.ch), or e164Number (e.g. 004112681549)

(H.323 defines other alias types, but they are rarely used)

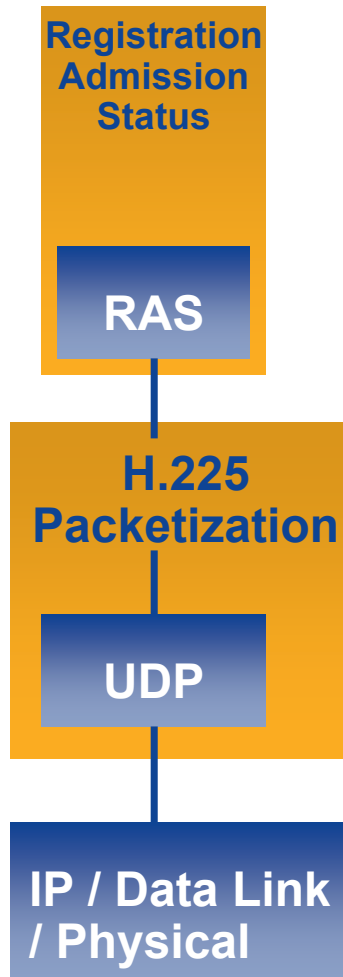
- **Admission control / bandwidth control**
 - Every call within the zone gets authorized by the GK
 - \mathcal{A} admission requests (destination address, bandwidth) to GK
- **Call control \mathcal{A} e.g. call transfer, call forwarding busy**
 - direct call signaling/control GK routed call signaling/control



H.323 Protocols Overview

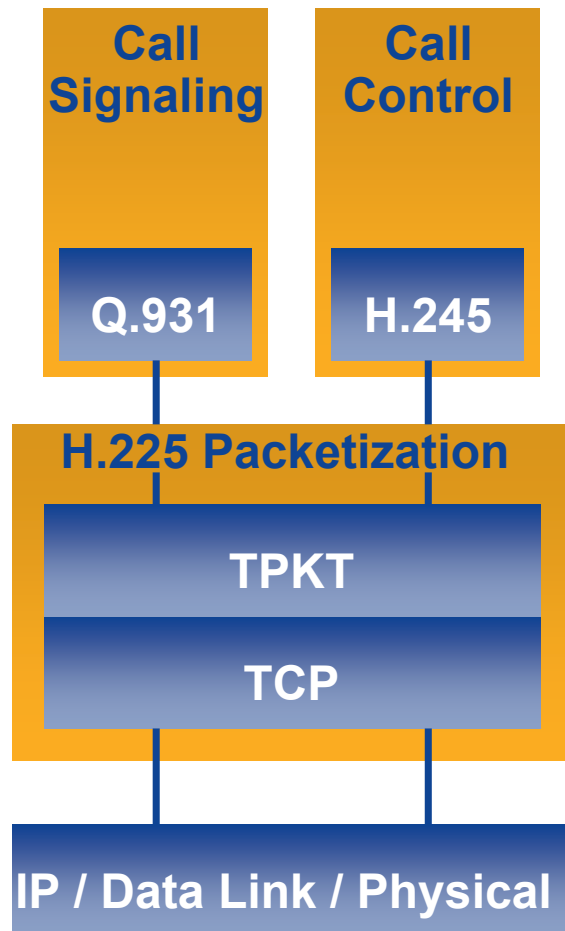


H.323 Protocols Overview: RAS



- Specified in H.225
- RAS messages are encoded using ASN.1
- RAS messages are used for:
 - Gatekeeper discovery (often done manually)
 - Gatekeeper registration
 - Name resolution (H.323 alias \neq IP Address)
 - Admission control
 - Bandwidth control
 - Status requests
- UDP, ports 1718 (GK discovery) / 1719

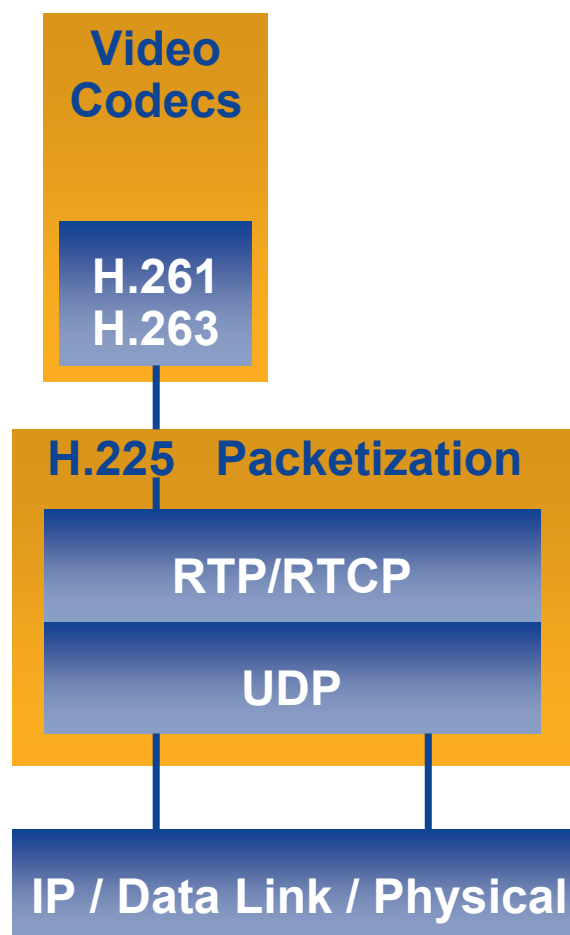
H.323 Protocols Overview: Q.931, H.245



- **Q.931**
 - call signaling protocol used in the *ISDN D-Channel*
 - Encoded in ASN.1
 - H.323 specific data in Q.931 UU-IE¹
- **H.245**
 - Master/slave determination
 - Capability exchange
 - Management of media and data streams
 - Encoded in ASN.1
- **TPKT**
 - delimit individual messages within the TCP stream (type, length header)
- **Q.931: TCP, port 1720**
- **H.245: TCP, dynamic port (>1024)**

¹ Q.931 UU-IE: User-User Information-Element: Optional element of a Q.931 packet, that includes application specific data (not defined by Q.931)

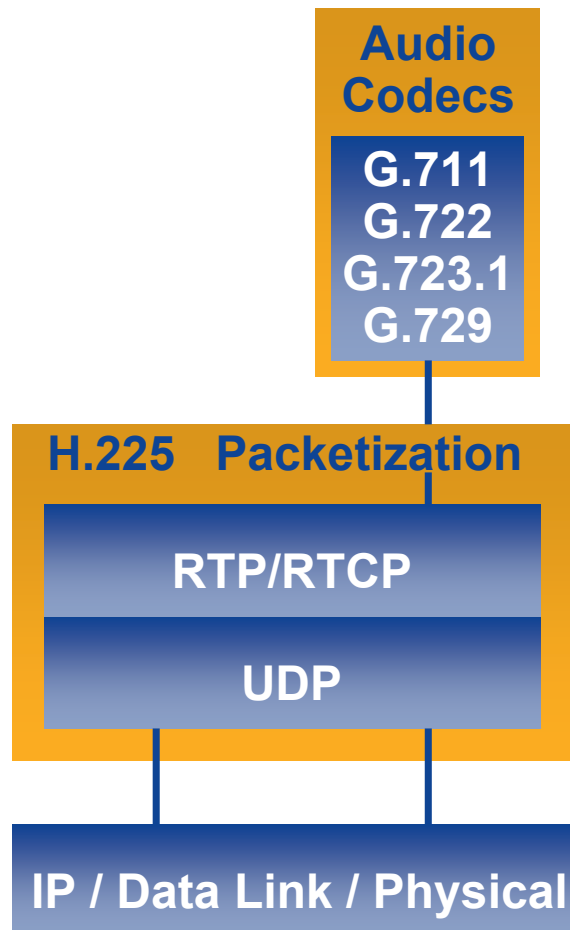
H.323 Protocols Overview: Video Codecs



- H.261
 - *Intra-frame* image compression (similar to JPEG, I-Frames) and *inter-frame* motion compensation (P-Frames)
 - QCIF, CIF¹ (optional)
- H.263
 - successor of H.261
 - same video quality as H.261 but lower bitrate
 - SQCIF, QCIF, CIF, 4CIF, 16CIF
- Future: H.264 = MPEG-4
- Real Time Transport Protocol (RTP)
 - IETF RFC 1889
 - end-to-end network transport function
 - payload type, sequence number, timestamp
- UDP, dynamic port (>1024)
(RTP: even, RTCP: odd numbered port)

¹ Common Intermediate Format (CIF): 352 x 288 pixels, QCIF = 1/4 CIF, SQCIF = 1/4 QCIF, 4CIF = 4 x CIF, 16CIF = 16 x CIF

H.323 Protocols Overview: Audio Codecs



- **G.711**
 - PCM, 64 kbits, voice quality: good
- **G.722**
 - 16 kbits, voice quality: low
- **G.723.1**
 - 5.3 kbits, voice quality: low
- **G.729**
 - 8 kbits, voice quality: good
- **RTP Control Protocol (RTCP)**
 - quality feedback, RTP session control
- **UDP, dynamic port (>1024)**
(RTP: even, RTCP: odd numbered port)

H.323 Protocols Overview: T.120



- **T.120**
 - Used for “data conferencing”
 - White board, image sharing (T.127)
 - File transfer (T.128)
 - Text chat (T.134)
 - Application sharing (non standardized)
- **TCP, port 1503**

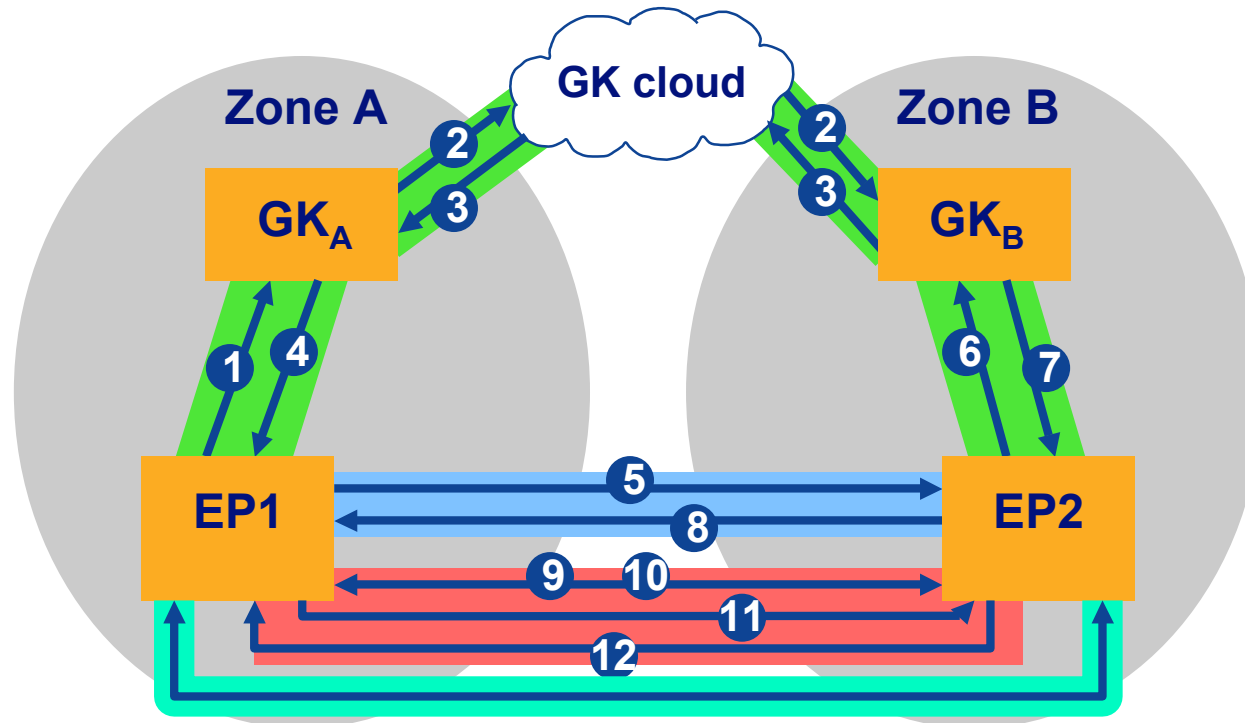
- **There are three phases to establish a call:**
 - **Phase A: GK Communication**
(admission, address translation)
 - **Phase B: Call Signaling**
(SETUP, ALERTING, CONNECT)
 - **Phase C: Call Control**
(Capability exchange,
open/close media streams)

Call signaling = connection level signaling
Call control = application level signaling

H.323 Call Establishment (direct call signaling)

Phase C: Call Control

Transmit and receive
media streams



- 1 AdmissionRequest (ARQ)
- 2 LocationRequest (LRQ)
- 3 LocationConfirm (LCF)
- 4 AdmissionConfirm (ACF)
- 5 Q.931 Setup
- 6 AdmissionRequest (ARQ)
- 7 AdmissionConfirm (ACF)
- 8 Q.931 Connect
- 9 Capability Exchange
- 10 Master/Slave
- 11 OpenLogicalChannel
- 12 OpenLogicalChannelAck

- RAS channel: UDP, port 1719 (H.225)
- Call Signaling channel: TCP, port 1720 (Q.931)
- Call Control channel: TCP, dynamic port >1024 (H.245)
- Audio/Video streams: UDP, dynamic port >1024 (RTP/RTCP)

- <http://www.packetizer.com/iptel/h323/>
(excellent H.323 Info Page, all standards for free!)
- <http://www.itu.int/rec/recommendation.asp?type=products&parent=T-REC-h>
(H series standards from the ITU)
- <http://www.openh323.org/>
(An H.323 open source implementation project)
- <http://www.iec.org/online/tutorials/h323/index.html>
(H.323 tutorial from Intel)