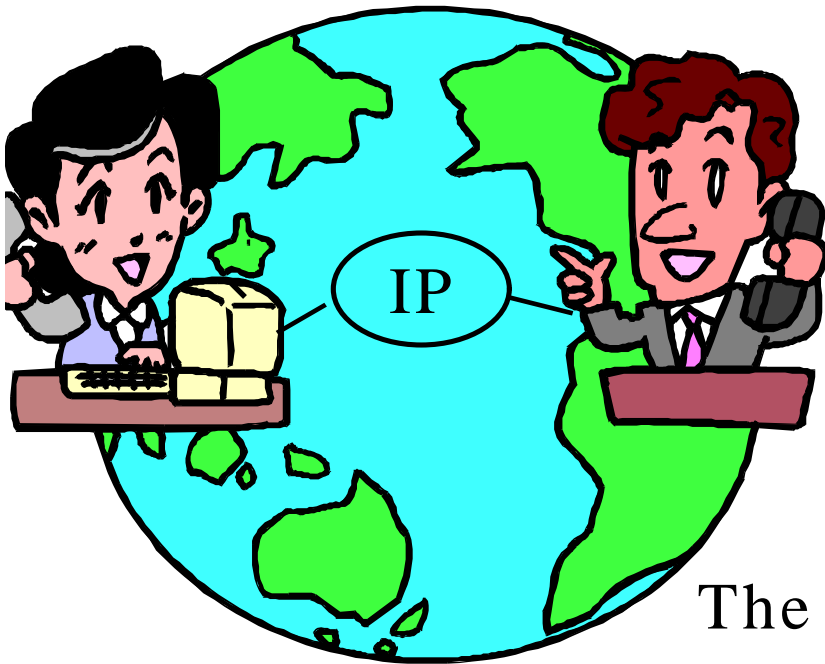

Voice over IP: Issues and Challenges



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Voice over IP: Why?

Sample Products and Services

3 Technical Issues

4 Other Issues

Protocols

H.323 Standard

Market

International VOIP calls could cost 1/5th of normal rates \Rightarrow Big share of \$18B US to foreign calls.

\$15B within Europe.

500,000 IP telephony users at the end of 1995.

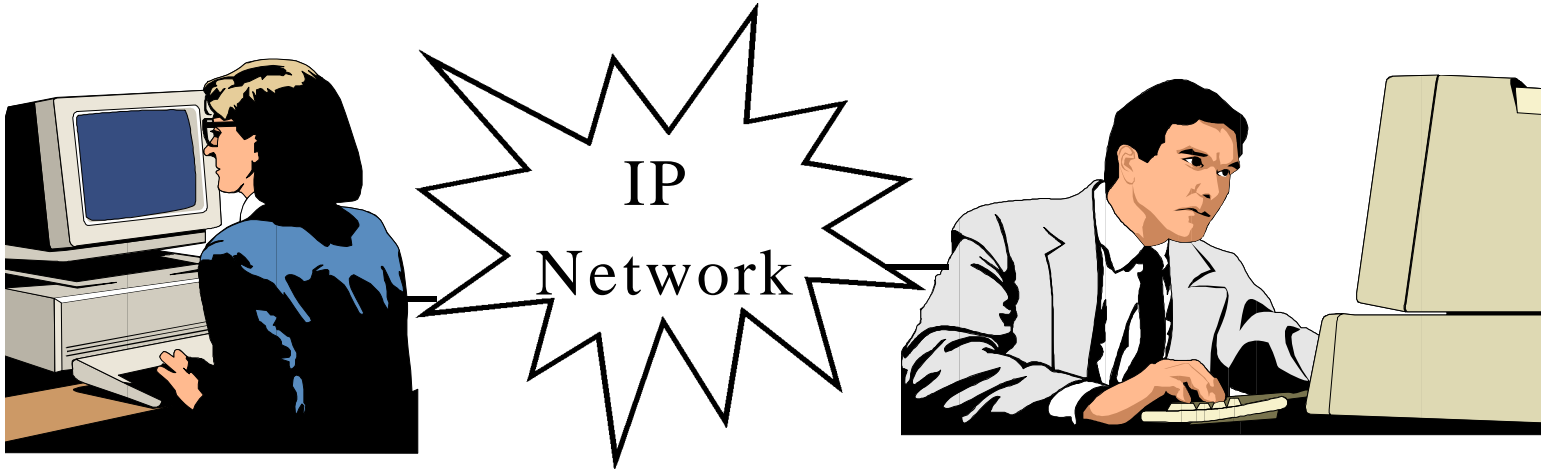
5% of all voice calls on IP/Internet by 2000

\Rightarrow 10M users and \$500M in VOIP product sales in 1999 [IDC]

US VOIP service will grow from \$30M in 1998 to \$2B in 2004 [Forester Research]

\$2B in 2001 and \$16B by 2004 [Frost & Sullivan]

Scenario 1: PC to PC

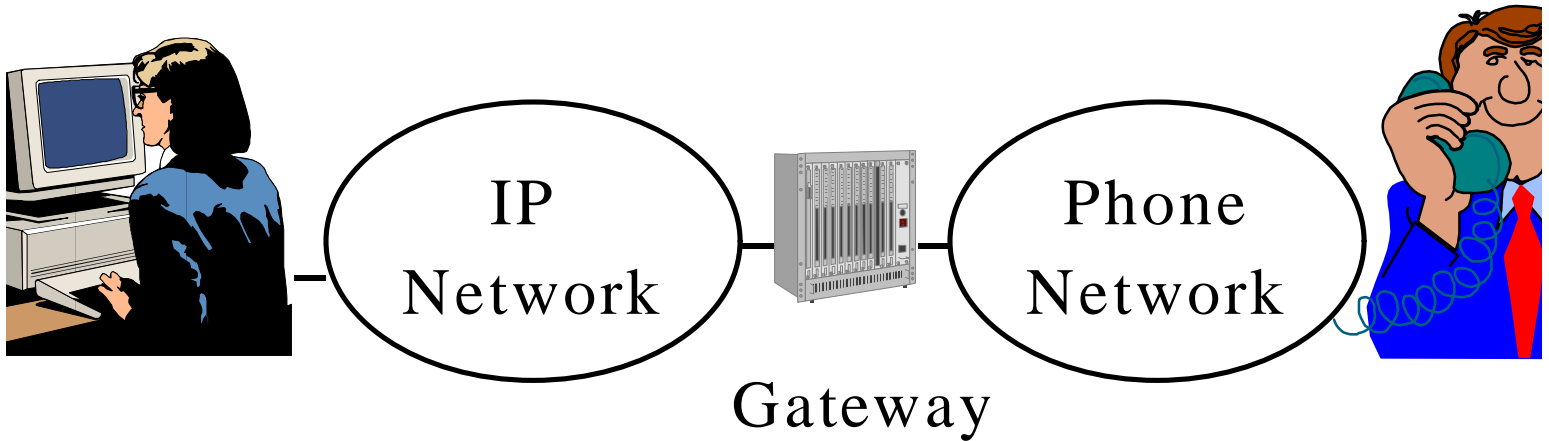


Need a PC with sound card

IP Telephony software: Cuseeme, Internet Phone, ..

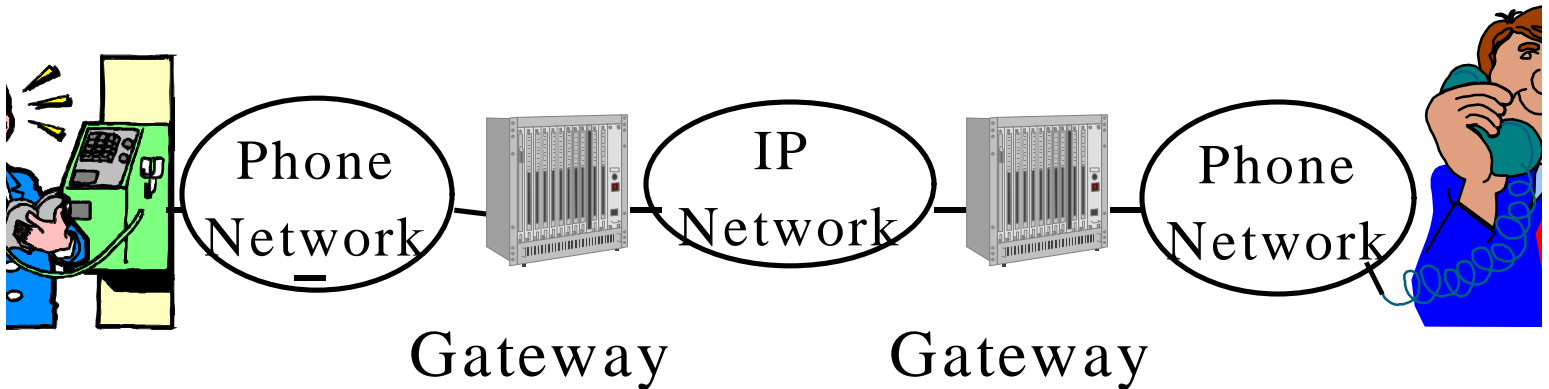
Video optional

Scenario 2: PC to Phone



Need a gateway that connects IP network to phone network (Router to PBX)

Scenario 3: Phone to Phone

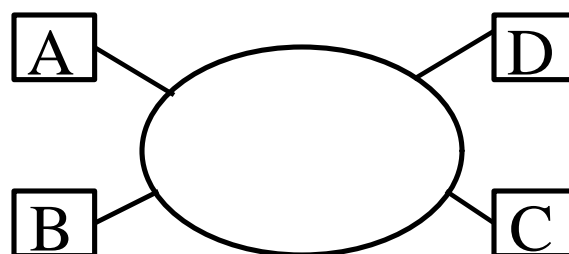
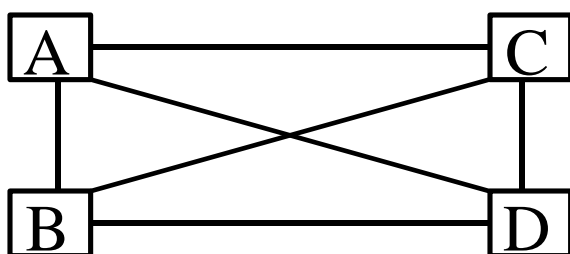


Need more gateways that connect IP network to phone networks

The IP network could be dedicated intra-net or the internet.

The phone networks could be intra-company PBXs or the carrier switches

Advantages



Private voice networks require $n(n-1)$ access links.

Private data networks require only n access links.

Voice has per-minute distance sensitive charge

Data has flat time-insensitive distance-insensitive charge

Easy alternate routing \Rightarrow More reliability

No 64kbps bandwidth limitation

\Rightarrow Easy to provide high-fidelity voice

Applications

Any voice communication where PC is already used

- Document conferencing
- Helpdesk access
- On-line order placement

international callbacks

many operators use voice over frame relay)

intranet telephony

internet fax

Sample Products

LocalTec Internet Phone: PC to PC.

Microsoft NetMeeting: PC to PC. Free.

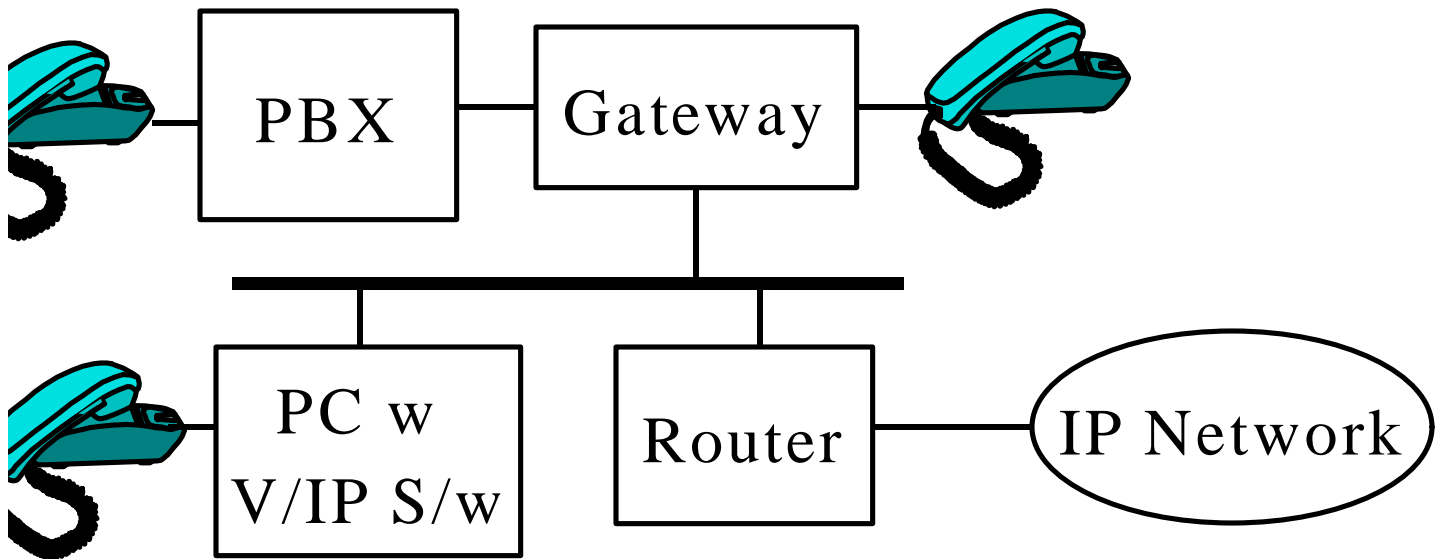
Internet PhoneJACK: ISA card to connect a standard phone to PC. Works with NetMeeting, InternetPhone, etc. Provides compression.

Internet LineJACK: Single-line gateway.

Micom V/IP Family:

- Analog and digital voice interface cards
- PC and/or gateway

Products (Cont)



○ Features:

- Compression
- Phone number to IP address translation.
- Supports RSVP.
- Limits number of calls.

Products (Cont)

LocalTec Internet Telephony Gateway:

- Similar to Micom V/IP
- Interactive voice response system for problem reporting
- Allows WWW plug in
- Can monitor other gateways and use alternate routes including PSTN
- Sold to Telecom Finland. New Zealand Telecom.

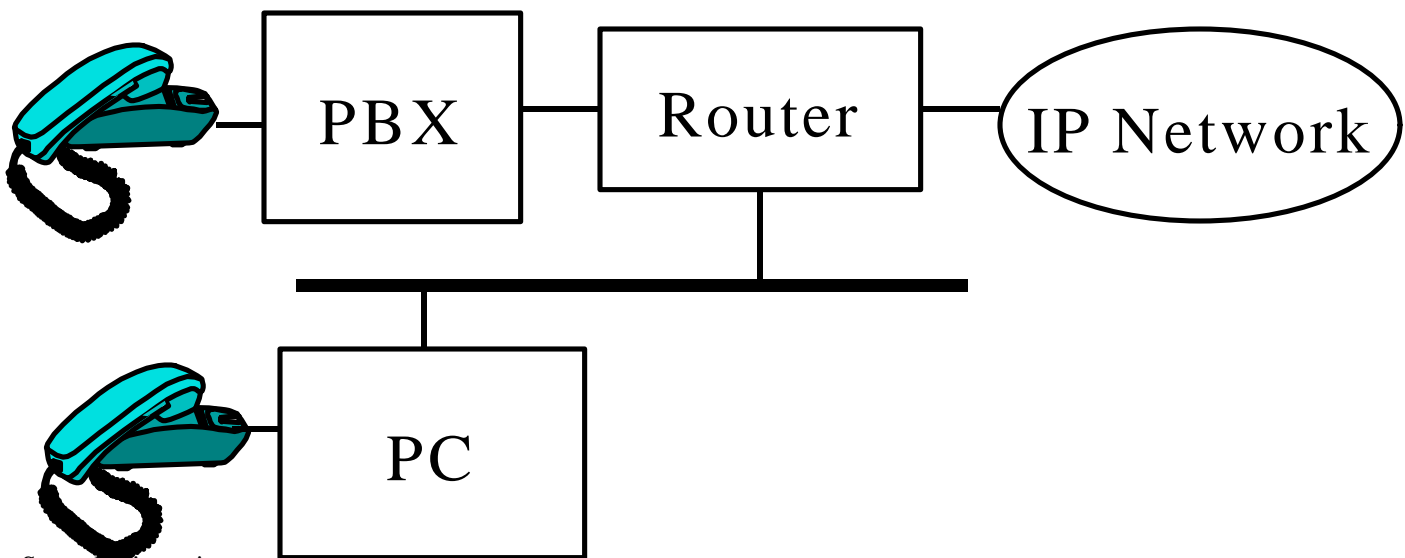
Lucent's Internet Telephony Server: Gateway|

Lucent PathStar Access Server

Products (Cont)

CISCO 2600 Routers: Voice interface cards (VICs)
Reduces one hop.

Baynetworks, 3COM, and other router vendors have
announced product plans



Sample Services

DT Corporation offers Net2Phone, Carrier2Phone, Phone2Phone services.

Global Exchange Carrier offers international calls using VocalTec InternetPhone s/w and gateways

Qwest offers 7.5¢/min VOIP Q.talk service in 16 cities.

TXC provides infrastructure and management to Internet Telephone Service Providers (ITSPs)

America On-line offers 9¢/min service.

AT&T announced 7.5¢/min VOIP trials in 9 US cities

Services (Cont)

Other trials: USA Global link, Delta 3, WorldCom, MCI, U.S. West, Bell Atlantic, Sprint, AT&T/Japan, KDD/Japan, Dacom/Korea, Deutsche Telekom in Germany, France Telecom, Telecom Finland, and New Zealand Telecom.

Level 3 is building a nation wide IP network for telephony.

Bell Canada has formed 'Emergis' division.

Bellcore has formed 'Soliant Internet Systems' unit

Bell Labs has formed 'Elemedia' division

Technical Issues

Large Delay

- Normal Phone: 10 ms/km \Rightarrow 30 ms coast-to-coast
- G.729: 10 ms to serialize the frame + 5 ms look ahead + 10 ms computation = 25 ms one way algorithmic delay
- G.723.1 = 100 ms one-way algorithmic delay
- Jitter buffer = 40-60 ms
- Poor implementations \Rightarrow 400 ms in the PC
- In a survey, 77% users found delay unacceptable.

Technical Issues (Cont)

Delay Jitter: Need priority for voice packets.
Shorter packets? IP precedence (TOS) field.

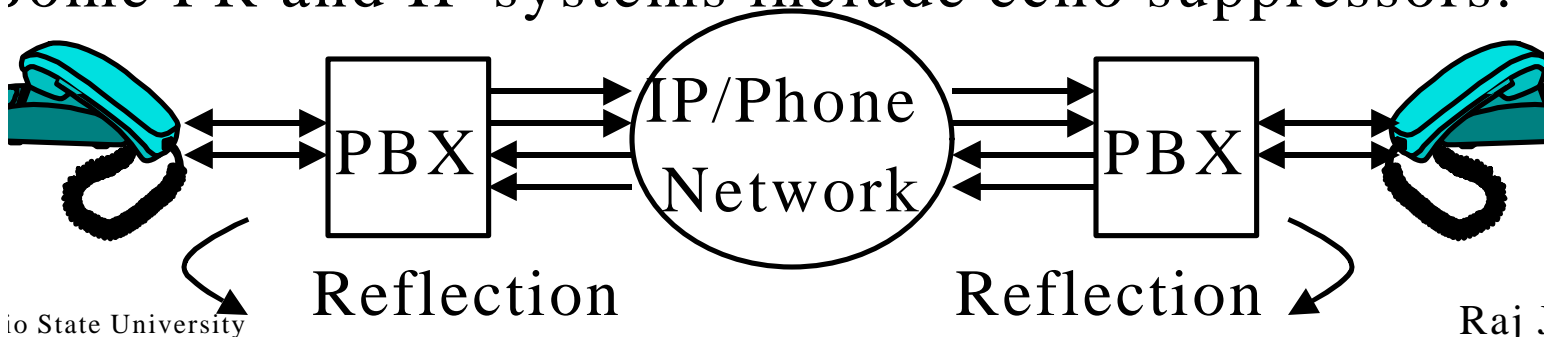
Frame length: 9 kB at 64 kbps = 1.125 s

Smaller MTU \Rightarrow Fragment large packets

Lost Packets: Replace lost packets by silence,
extrapolate previous waveform

Echo cancellation: 2-wire to 4-wire.

Some FR and IP systems include echo suppressors.



Technical Issues (Cont)

Silence suppression

Address translation: Phone # to IP. Directory servers

Telephony signaling: Different PBXs may use different signaling methods.

Bandwidth Reservations: Need RSVP.

Multiplexing: Subchannel multiplexing

⇒ Multiple voice calls in one packet.

Security: Firewalls may not allow incoming IP traf

Insecurity of internet

Voice compression: Load reduction

Other Issues

Per-minute distance-sensitive charge vs flat time-insensitive distance-insensitive charge

Video requires a bulk of bits but costs little.

Voice is expensive. On IP, bits are bits.

National regulations and government monopolies

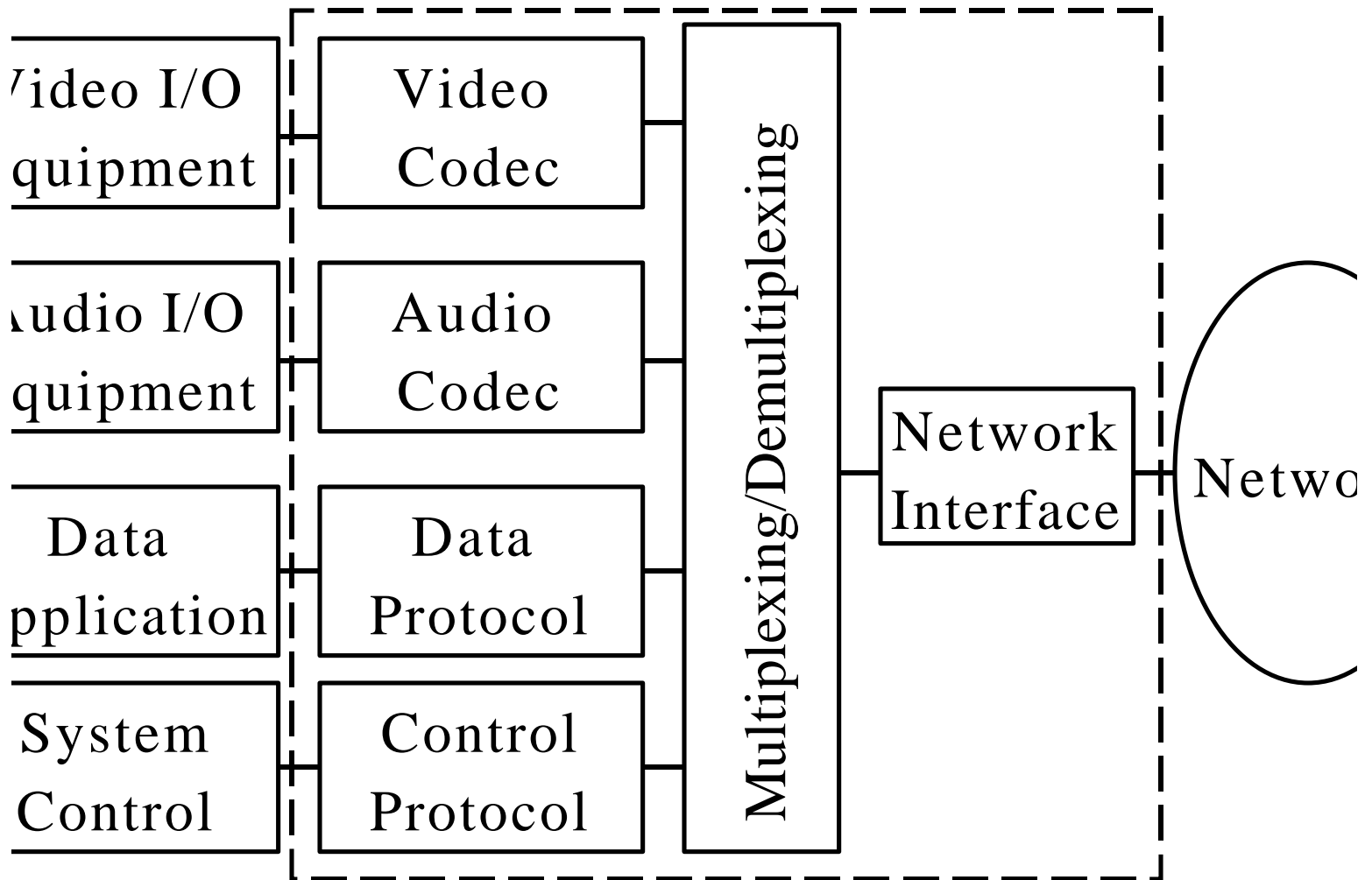
⇒ Many countries forbid voice over IP

In Hungary, Portugal, etc., it is illegal to access a website with VOIP s/w. In USA, Association of

Telecommunications Carriers (ACTA) petitioned FCC to levy universal access charges in ISPs

Modem traffic can't get more than 2400 bps.

telephony/Conferencing System



Conferencing Standards

| | | | | | |
|---------------------|---------------------------|---------------------------|---------------------------|--|-------------------|
| Network | ISDN | ATM | PSTN | LAN | POTs |
| Conf. Std. | H.320 | H.321 | H.322 | H.323 V1/V2 | H.324 |
| Year | 1990 | 1995 | 1995 | 1996/1998 | 1996 |
| Audio Codec | G.711, G.722, G.728 | G.711, G.722, G.728 | G.711, G.722, G.728 | G.711, G.722, G.723.1, G.728, G.729 | G.723.1, G.729 |
| Audio Rates Kbps | 64, 48-64 | 64, 48-64, 16 | 64, 48-64, 16 | 64, 48-64, 16, 8, 5.3/6.3 | 8, 5.3/6.3 |
| Video Codec | H.261 | H.261, H.263 | H.261, H.263 | H.261 H.263 | H.261 H.263 |
| Data Sharing | T.120 | T.120 | T.120 | T.120 | T.120 |
| Control | H.230, H.242 | H.242 | H.242, H.230 | H.245 | H.245 |
| Multiplexing | H.221 | H.221 | H.221 | H.225.0 | H.223 |
| Signaling | Q.931 | Q.931 | Q.931 | Q.931 | - |

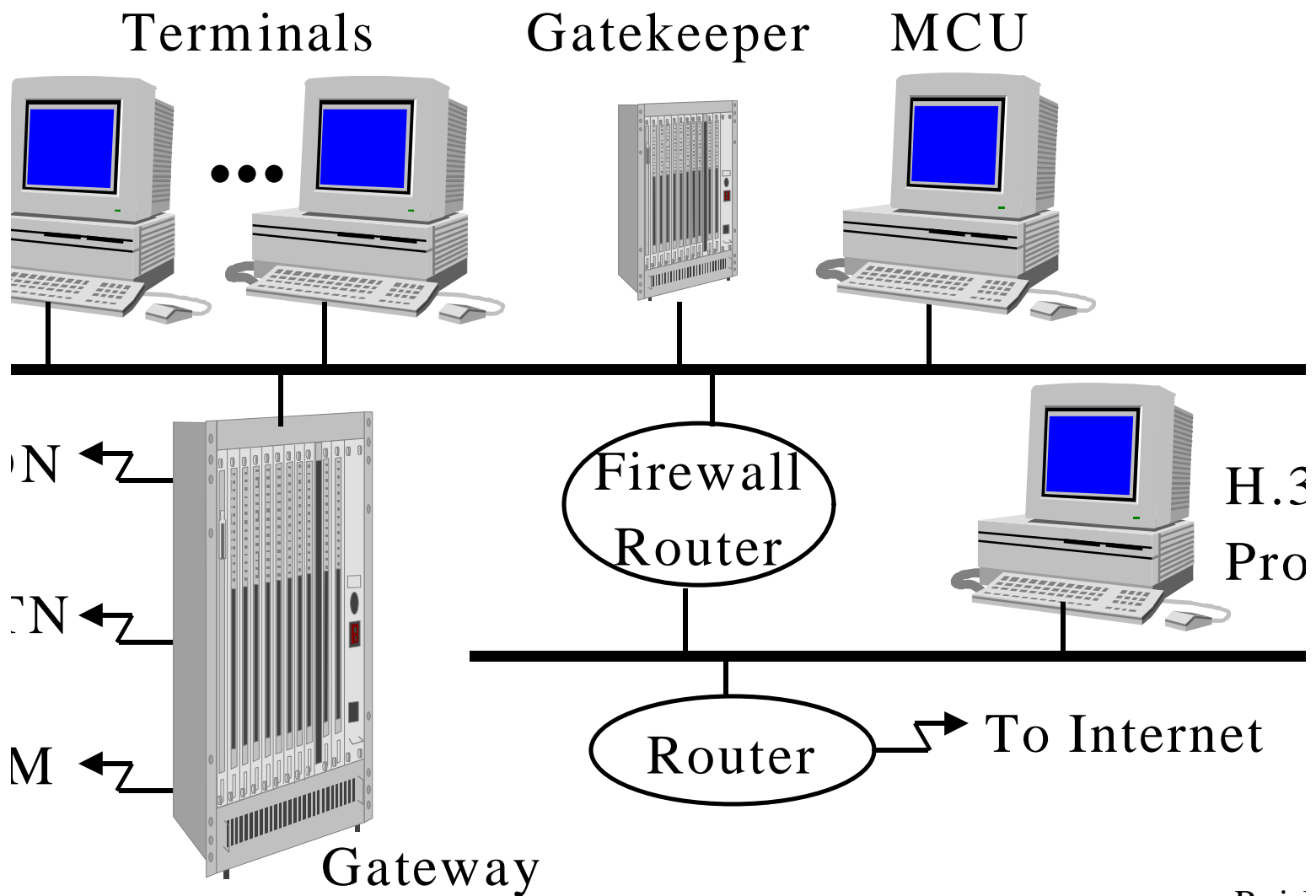
H.323 Protocols

Multimedia over LANs

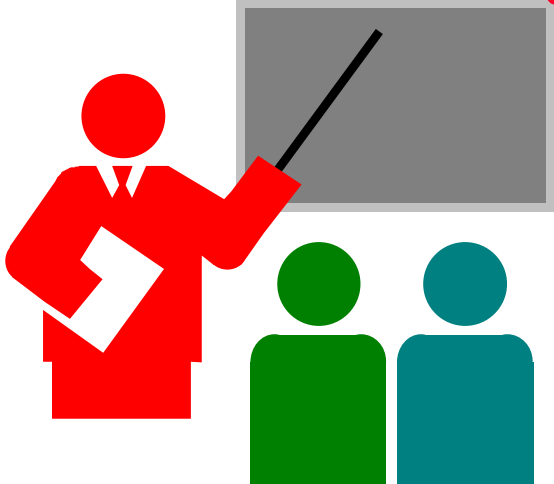
Provides component descriptions, signaling procedures, call control, system control, audio/video codecs, data protocols

| Video | Audio | Control and Management | | | Data | |
|-----------------------|---|------------------------|----------------|----------------------|------------------|------|
| H.261 H.263 | G.711, G.722, G.723.1, G.728, G.729 | RTCP | H.225.0 RAS | H.225.0 Signaling | H.245 Control | T.11 |
| RTP | | | X.224 Class 0 | | | T.11 |
| UDP | | TCP | | | T.11 | |
| Network (IP) | | | | | T.11 | |
| Datalink (IEEE 802.3) | | | | | | |

H.323 Components



Summary



Voice over IP products and services are being rolled out
deal for computer-based communications

IP needs QoS for acceptable quality

A number of working group at IETF are working on
RFC.323 provides interoperability

References

See

http://www.cis.ohio-state.edu/~jain/refs/ref_voip.htm

for a detailed list of references.

Thank You!

