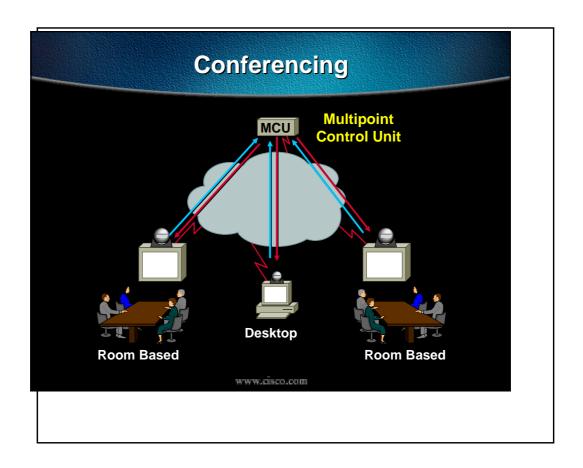
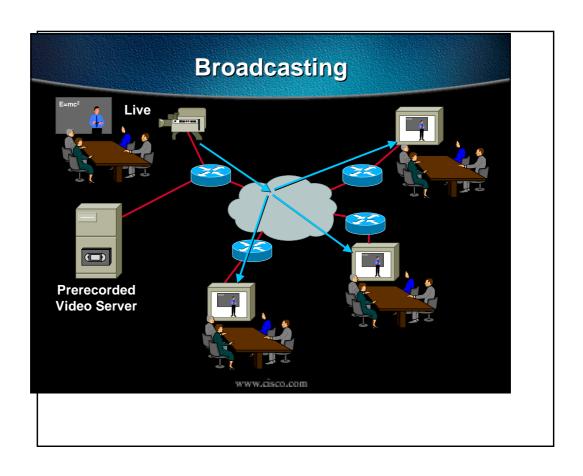


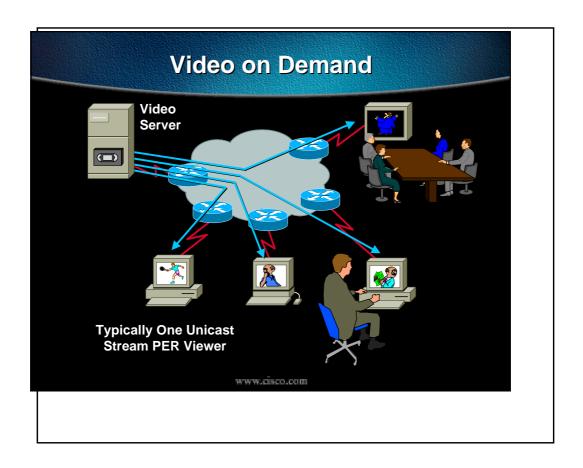


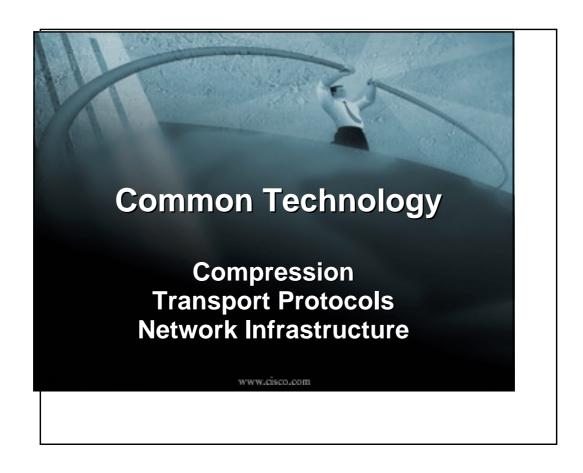
# Agenda • Application Classes • Common Technology • Protocols and Architectures • Network Planning

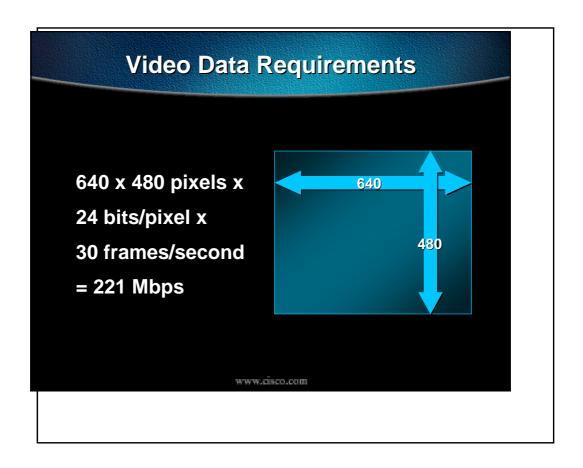












	Video Quality	Bandwidth (bits/sec)	Latency (ms)
<b>Motion JPEG</b>	Broadcast	10–26 M	<45
MPEG-2	Broadcast/HDTV	3–16 M	200–500
MPEG-1	VCR/Business	.5–1.5 M	200–500
H.261/H.263	Video Conf.	64 K–2 M	150–250
MPEG-4	Internet/Business	< 64 K-4 M	150–500
	_		

### **Audio Compression Standards**

Method	Source BW	Compress Ratio	Bandwidth Requirement
G.723.1	64 Kbps	10:1	6.4 Kbps
G.729	64 Kbps	8:1	8 Kbps
G.721	64 Kbps	2:1	32 Kbps
G.722	224 Kbps	3.5–4.6 : 1	48–64 Kbps
MPEG	706 Kbps	3–11 : 1	64–256 Kbps

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### **Real-Time Transport Protocols**

- Proprietary protocols for streaming
  - Real Networks, Microsoft, others (primarily unicast)
- IETF standards track protocol: RTP/RTCP

Conferencing (H.323) and streaming (Cisco IP/TV® and Apple QuickTime)

### RTP/RTCP

- Real-time transport protocol
   Carries the data (media)
- Real-time transport control protocol
   Carries control info among all session members
- Specified in RFC 1889

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### **RTP/RTCP Features**

- RTP and RTCP both based on UDP
  - Low delay is more important than 100% reliability
- Designed from the start to operate in multicast as well as unicast modes

### **UDP + RTP = Transport Layer**

- UDP provides checksum and length
- RTP end-to-end network transport functions:
   Payload type identification—voice, video, compression
   Sequence numbering

Time stamping

4bytes | CC | M | Payload | Sequence Number |
4bytes | RTP Timestamp |
4bytes | Synchronization Source (SSRC) ID

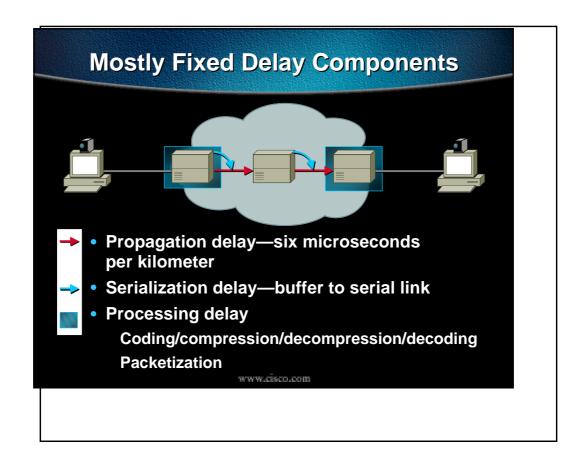
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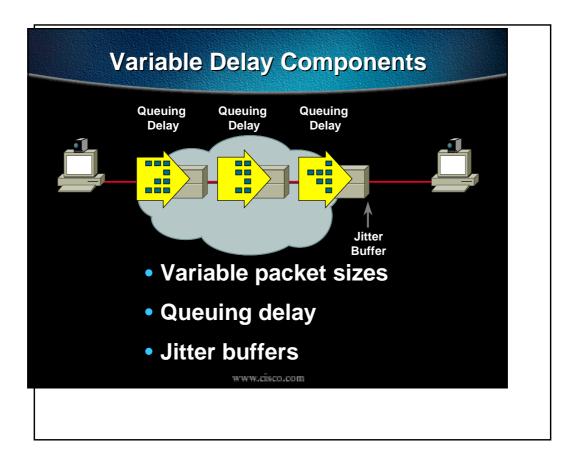
### **RTCP Functions**

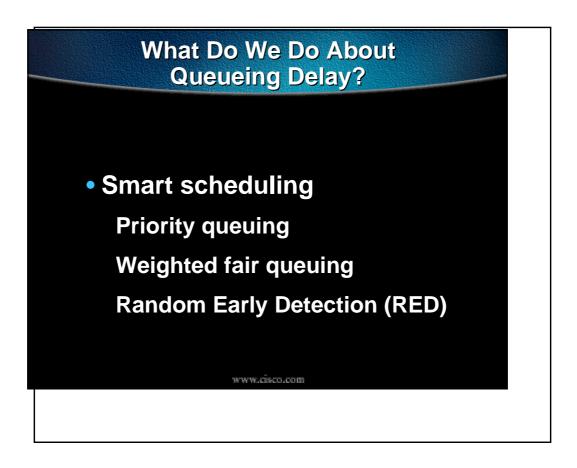
- Provides feedback on distribution quality (especially loss rate)
- Provides a uniform time reference for synchronization between separate RTP sessions
- Identifies all senders and receivers

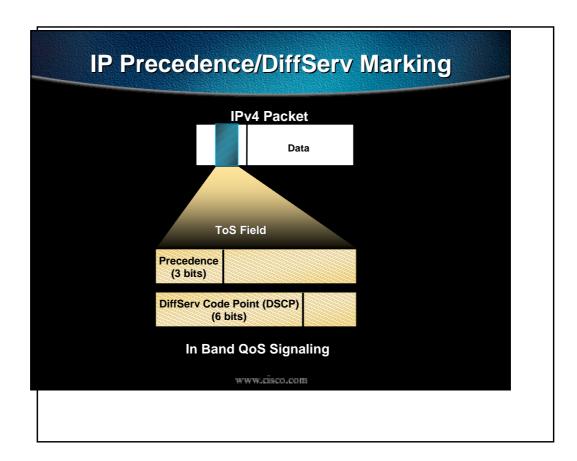
# What Does Video Want from the Network?

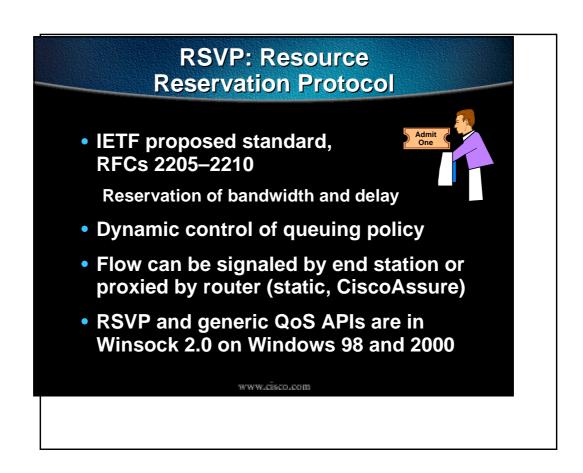
- Sufficient bandwidth
- Low packet loss (avoid congestion)
- Low delay (jitter increases max delay)
- Multicast delivery for broadcast or large conferences



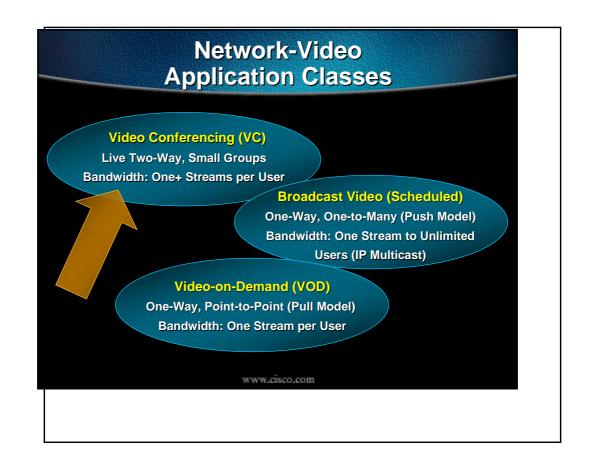






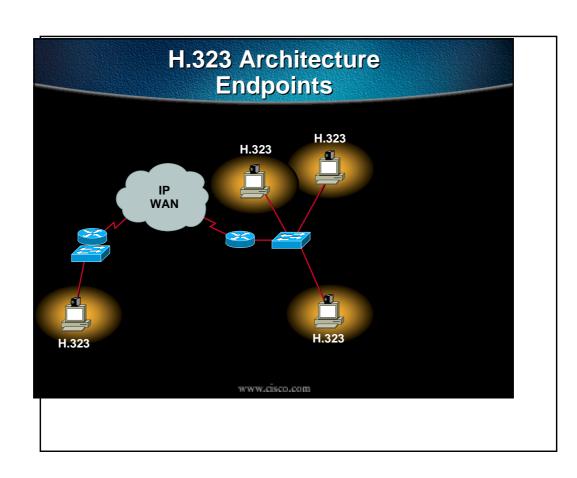


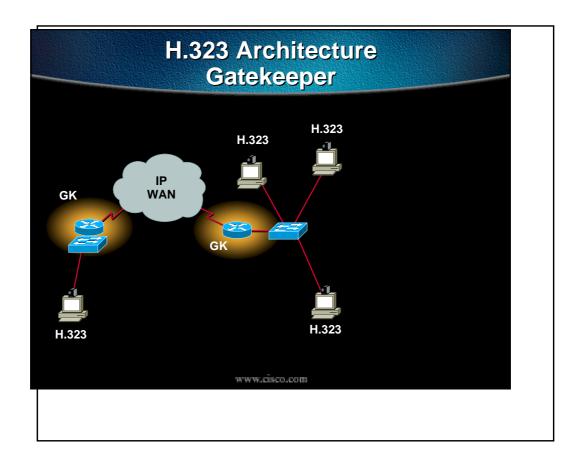


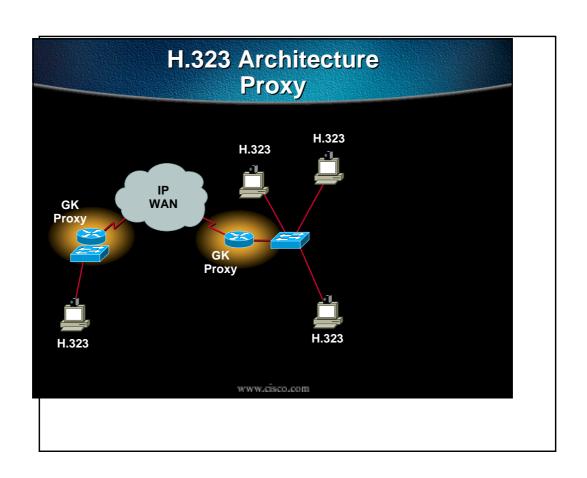


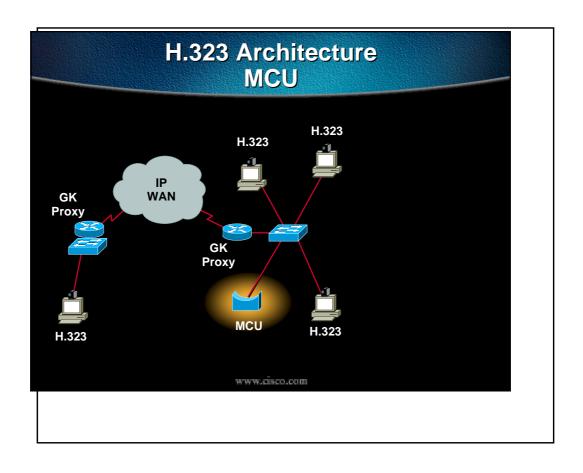
# **IP Conferencing Protocols**

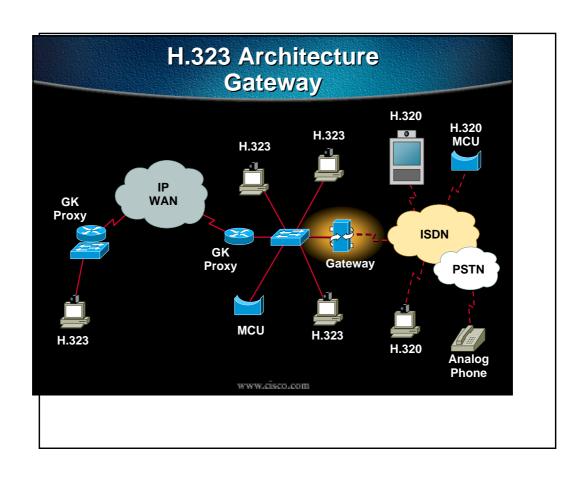
- H.323
  - ITU standard protocol
    Evolved from H.320 ISDN standard
- SIP—Session Initiation Protocol IETF proposed standard
   Simpler, but less mature, than H.323

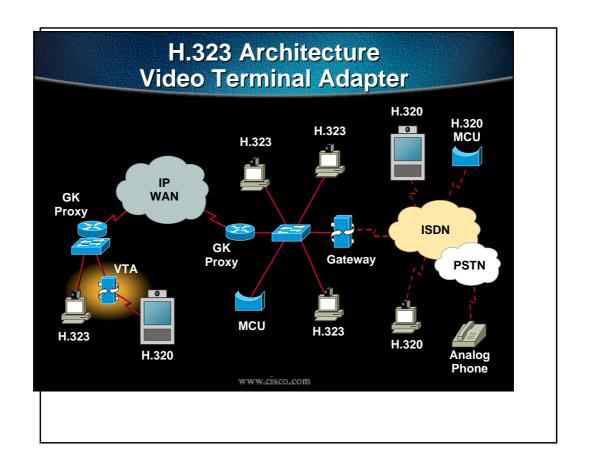


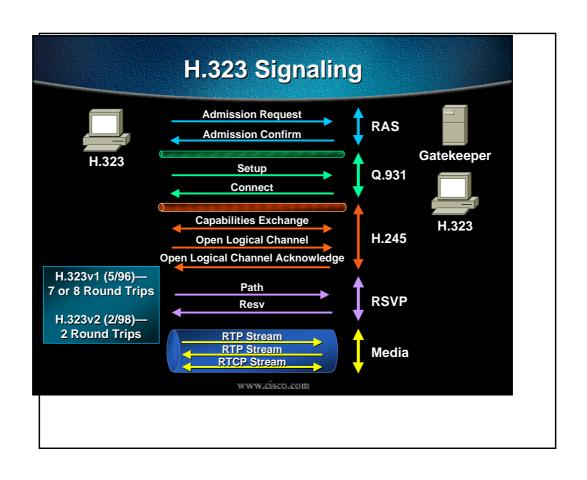












### SIP: Session Initiation Protocol

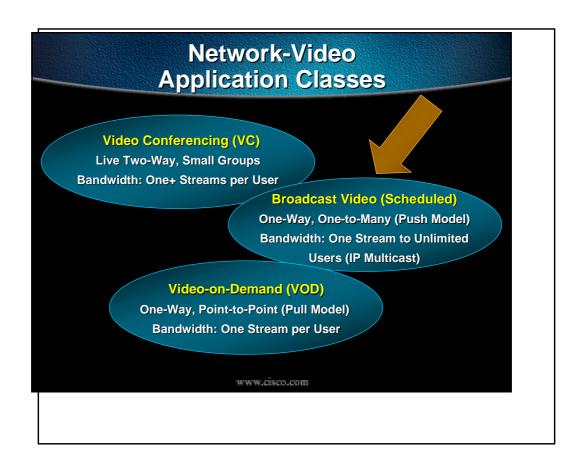
- Lightweight signaling protocol, in part a reaction to complexity of H.323
- IETF Proposed Standard RFC 2543, March 1999
- SIP bake-offs April 1999, December 1999 with 26 organizations

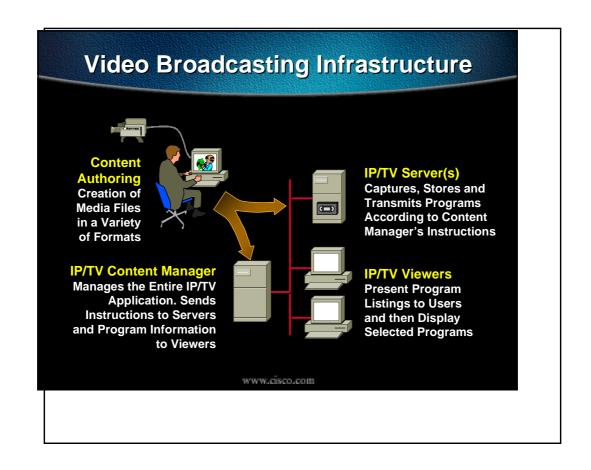
Clients, proxy/redirect servers, PSTN gateways...

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### **SIP Features**

- Text-based in style of HTTP
  - → INVITE user@host.net
  - **←**200 OK
- Post-dial delay = 1.5 RTT
- Allows stateless servers for scalability
- Uses SDP to describe call parameters





### **Streaming File Formats**

- Real Networks—Proprietary formats including multirate video
- Microsoft—AVI, ASF (advanced streaming format)
- Apple QuickTime—Basis for MPEG-4
- MPEG-1/2 Elementary and Systems Streams, "MP3" audio
- RTP file formats

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### **Choosing a Format**

- Depends upon what delivery system will be used
- Driven by what authoring systems produce
   Editability
- Need multirate streaming?
   Want scalable encoding or multistream file format

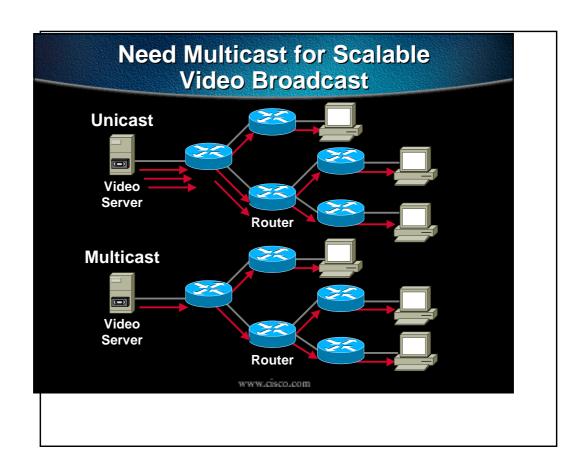
# **Control Protocols** for Broadcasting

- Much simpler than conferencing
- Web interface

Link invokes player with necessary session parameters

Announce/subscribe protocols

Session parameters distributed to potential receivers Examples: SMIL (Real), SDP (IP/TV)



### **IP vs. Application Multicast**

Network-level IP multicast:

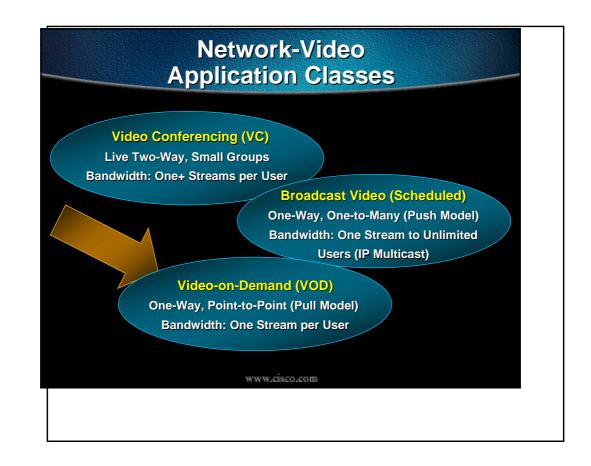
Builds a dynamic distribution tree using standard, multivendor protocols

Automatically reconfigures distribution tree as receivers come and go

Application-specific schemes:

"Reflectors" and "splitters" form a rigid distribution tree (static routing)

Need new agents for each application

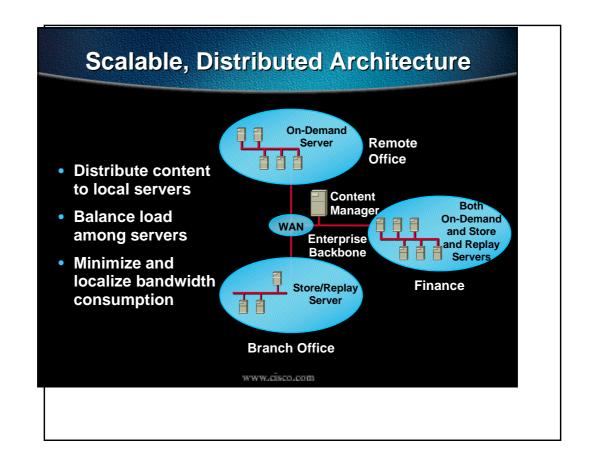


### What Is Video on Demand?

- Architecture like video broadcasting:
   Servers, viewers, content management
- But:

Video available "immediately" upon request

- **→** Unicast stream per user
- Server and network capacity scaling is an issue



### **VOD Control Protocols**

Access via:

Web browser and plug-in Application program listing

- Server session control protocol:
   Usually TCP because point-to-point
   Provides asset selection, VCR controls
- Proprietary protocol or RTSP

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# RTSP: Real-Time Streaming Protocol

- IETF Proposed Standard RFC 2326
- Text-based in style of HTTP
  - → PLAY rtsp://foo.com/bar.file RTSP/1.0
  - **←RTSP/1.0 200 OK**
- Uses SDP to describe media
- Uses RTP for media transport



# **Domains for Scalability**

- Conferencing applications
   Gatekeeper zones
   Proxies and firewalls at boundaries
- Streaming applications
   Multicast administrative scopes
   VOD clusters

### **The Video Quality Conundrum**

- Video quality is directly related to bandwidth
- Bandwidth is usually a fixed, limited resource
- All unicast solutions force a compromise between the number of users supported and video quality

The more users, the lower the bandwidth per user must be, thereby lowering the video quality

Delay also impacts quality for interactive video

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### **Bandwidth and Load**

Video Conferencing	128 Kbps 384 Kbps	100 pps 100 pps
MPEG-1/MPEG-2	1 Mbps 6 Mbps	85 pps 500 pps
Video on Demand	128 Kbps 512 Kbps 1 Mbps	11 pps 44 pps 88 pps

# **Choosing Quality/Bandwidth**

- Know your target application
   Example: Enterprise vs. Internet
- But don't aim too low28.8 video is a teaser, not a real solution
- Plan to increase network capacity
   Compression won't improve 10X







IP Frames or ATM Cells?					
	IP	ATM			
Transport	Ethernet(s) FDDI ATM	ATM Only			
QoS	RSVP IP Precedence	VBR, CBR			
Bandwidth	Assured thru Routers	Guaranteed by Switches			
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# **IP Quality of Service**

- IP QoS maps onto any Layer 2 QoS ATM 802.1p
- Routers can do QoS for dumb links
   Task is same for packets or cells

### **Deploying IP Multicast**

Use administrative scoping, RFC2365

Keeps traffic where you want it, e.g. multiple scopes for different bandwidths

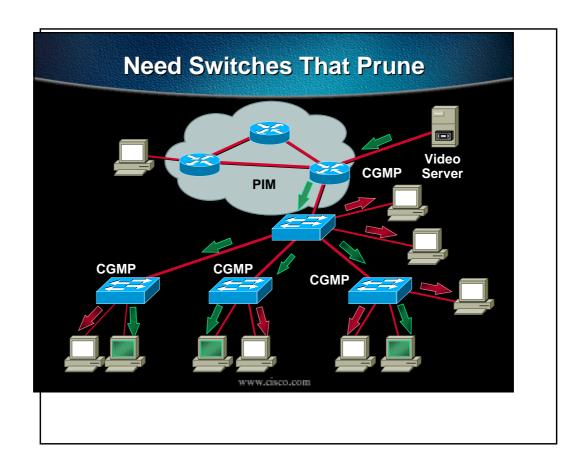
Allows local multicast address allocation and address reuse

Rendezvous Point (RP) placement
 Consider splitting load on multiple RPs

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### **Choose the Right Equipment**

- Use devices that can handle high levels of multicast traffic well:
   Replication at high-traffic levels
   Large numbers of groups and sources
- Some potential problem areas:
   ATM LANE broadcast unknown server
   Some DSL devices
- Recommend switches, not hubs



### **Summary**

- There are many different applications for packet video
- Supported by a well established body of technology
- Protocols and architectures have been developed to meet the needs of each class of applications
- Network infrastructure components are ready to be deployed

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