



Computer Networks

Multimedia and Multicast

Outline

- ◆ Multimedia Overview
- ◆ Receiver-Driven Layered Multicast
- ◆ UDP Sockets (coming soon)
- ◆ IP Multicast (coming soon)
- ◆ Misc (coming soon)

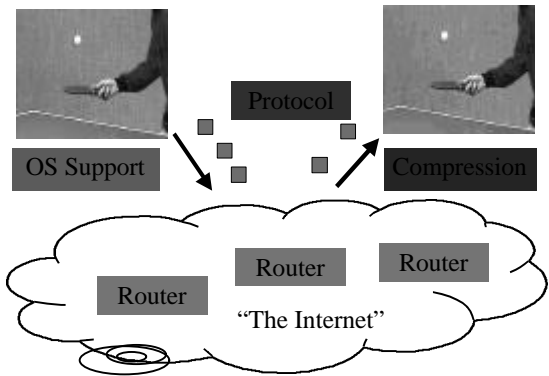


Multimedia Overview

- ◆ Talking about *continuous* media
 - RealAudio, RealVideo, Internet Phone
- ◆ Typically thought of as high-bandwidth
 - raw video 30 Mbps
 - but not necessarily true
 - ◆ compressed audio 8 Kbps
 - ◆ compressed video 2 Mbps
- ◆ New computer is "multimedia ready"
 - plenty of CPU power
 - special devices (MMX, video chips ...)
- ◆ So ... what's the problem?



Multimedia on the Internet

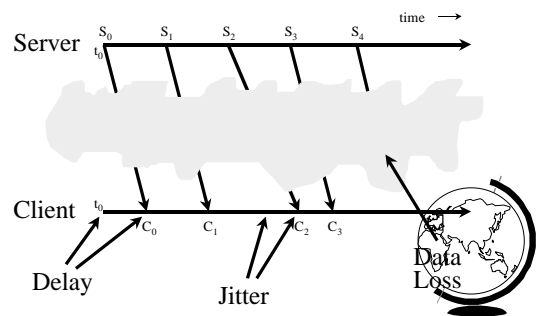


Multimedia Overview

- ◆ Today, just about every new computer is "multimedia ready"
 - plenty of CPU power
 - special devices (MMX, video chips ...)
- ◆ So ... what's the problem?



Multimedia Performance



Internet Shortcomings

- ◆ Designed for “text-based” applications
 - *without* strict timing constraints
 - *with* strict loss constraints
- ◆ “Bursty” traffic
 - high variance in delay
 - periods of heavy packet loss
- ◆ Limited network protocols for applications



Internet Protocols

- | | |
|---|--|
| <ul style="list-style-type: none"> ◆ TCP <ul style="list-style-type: none"> – delivers every byte <ul style="list-style-type: none"> ◆ unbounded delay! – stream semantics – fixed flow control – unicast – ... big bleah! | <ul style="list-style-type: none"> ◆ UDP <ul style="list-style-type: none"> – “best-effort” delivery <ul style="list-style-type: none"> ◆ unbounded loss! – packet semantics – no flow control – <i>multicast</i> add-on – ... bleah! |
|---|--|

“Sigh. I guess I’ll use UDP since it is better than TCP. Or ... not?”



The Internet Today

- ◆ Mostly TCP traffic
 - 96%: ftp, telnet, nntp, smtp... (tcplib’92)
- ◆ Optimized for TCP
 - “Thinner” OS protocol stacks
 - Vegas, Reno, Tahoe ...
- ◆ Punish “non-responsive” flows
 - UDP
 - RED, ECN

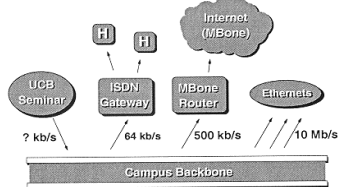


Receiver-driven Layered Multicast

Steven McCanne, Van Jacobson and Martin Vetterli

ACM SIGCOMM, Stanford CA, August 1996

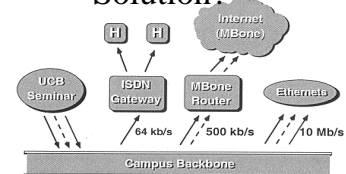
Problem



- Network heterogeneity
- One output to multiple users with varied capabilities
- Who decides the rate?
- What is the network capacity ?

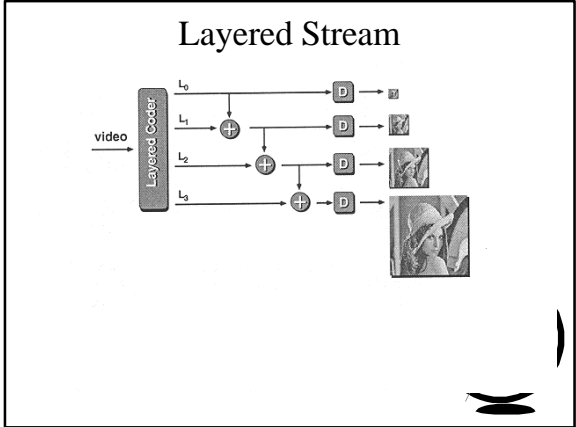


Solution?



- Multiple levels of quality across multiple network channels
- Receivers decide their own rates of reception
- Note, requires layered media streams



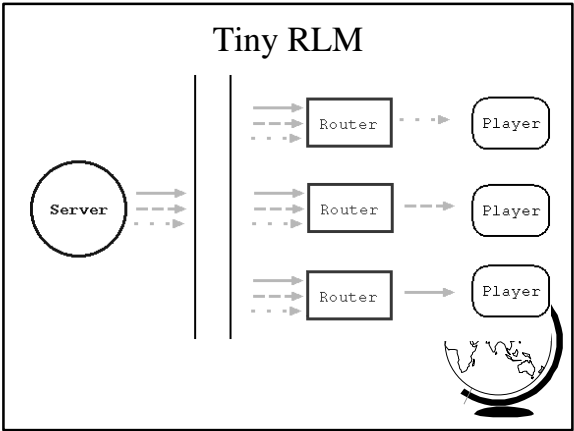


The RLM Protocol

- ◆ High level abstraction
 - on congestion, drop a layer
 - on spare capacity, add a layer
- ◆ Q: How does the receiver decide?
 - detection time
 - capacity inference

Event Sequence

- At a well-chosen time conduct a join experiment
- If congestion is experienced, leave the new group
- If no congestion, try to join next higher group



Tiny Movies

Taking a Walk

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Taking a Walk

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Taking a Walk

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- ◆ Text-based frames
- ◆ One frame per second
 - sleep! alarm! setitimer!