Computer Networked games

- Another form of multimedia traffic audio, video and interactive
- Slides courtesy Mark Claypool @ WPI



Game Types

- First Person Shooters
 - Doom, Quake, Counter-strike, ...
- Massive Multi-Player Online Role Playing
 - Everquest, Earth and Beyond, ...
- Real-Time Strategy
 - Warcraft, Starcraft ...
- Other
 - Misc not any genre above
 - Example: Diablo2, Racing, ...
 - Non-networked
 - Example: Thief
 - Multiplayer, but relaxed real time
 - Example: Chess, Bridge





Example of Unresponsiveness



Player is pressing left



Player is pressing up



Running back goes out of bounds! Player curses

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Why Does Latency Matter?



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How Much Does Latency Matter?





Depends upon precision of task

How Much Does Latency Matter?



Depends upon deadline of task

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Player Performance vs. Latency



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Latency Compensation - Time Warp

Server rolls-back (warps) to when action taken place on client



Disadvantage - user may notice when fix state inconsistency

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Computer Networks...

- Brief overview of computer networks and how it plays into multimedia
- Network types:
 - Circuit switched: e.g., telephone lines
 - Dedicated link. Have enough bandwidth for a telephone call whether we say anything on the line or not. This is good if we don't use any compression, wasteful if we use any compression (silence can be easily eliminated)

Packet switched: modern computer networks

- Data is packaged into packets which are individually routed across the network.
 - Slightly wasteful because each packet carries identity and destination information. Not as bad as sending (say) a 1000 byte packet as opposed to 1 byte packet
 - Packets compete for network resources leading to harder to predict networking

Packet switched connection models

TCP – reliable, in-order at most once delivery

- Eases developer effort to manage reliability
- Network retransmits data to ensure in-order delivery
 - Multimedia applications can either tolerate dropped packets or desire to stay abreast of live stream rather than wait for stragglers
- TCP traverses home routers/firewalls in the reverse direction (outside to inside) better than UDP, especially on cheap routers. Hence, TCP remains a viable option for games
- TCP internally buffers data to achieve efficiency, sending 1 byte packets might be collected to become a 1000 byte packet – unacceptable for multimedia (games)

UDP

- Data gram protocol no guarantees of order of delivery or reliability
 - Multimedia applications have full control over what to transmit and when. They also have control over dropped packets. For example, I frames might be retransmitted or forward corrected, lost B frames might be ignored etc.



Networks strive for fairness

- Assuming we want to support 'n' network flows over a network link of capacity 'm'
 - Approach 1: reserve equally, say n/m per flow or proportional to flow requirements
 - Wasteful if flow doesn't use a particular allocation
 - Sharing can help customers even though ISP get paid
 - Approach 2: Best effort
 - Each flow can be aggressive and pump as much as they possibly can. Network will experience congestion and drop packets. You will get the best of what can be supported
 - Each flow can be "nice". One form of niceness is TCP friendly. TCP internally probes the network bandwidth using additive increase multiplicative decrease (AIMD) scheme. It increases the amount of data slowly in additive fashion till it experiences congestion (packet drop). At that point, it drops quickly and tries again to calculate the bottle neck bandwidth in a distributed fashion

Multimedia traffic

- Isochronous traffic: somewhere between synchronous (constant traffic) and asynchronous (unpredictable). Sort of predictable, sort of bursty
 - Peak burst not necessarily equal to the network capacity (unlike for file transfer application which can use as much bandwidth as you can throw at it).
 - Traffic bursty and will interact with TCP traffic that are competing for resources



Microsoft Media

- Packets arrive at fairly regular intervals
- Large packets (up to 16 KB): Uses network fragmentation
 - Losing one fragment loses the entire packet
 - Fragments arrive back to back
 - can assist adaptation policy
- Lossy network adapts to a low quality stream
 - fragmentation makes this effect worse



Real

- Variable packet size (50-1500 bytes)
 - Packet arrivals almost regular
 - Packets sent closer to each other
- Packet size less than MTU
 - No network level fragmentation
- Lossy networks lower quality video



Quicktime

- Variable packet size
 - Packets sent in "clusters" burst and extended idle
 - Application level fragmentation?
- Packet size less than MTU
 - No network fragmentation
- Lossy networks lower quality video



Networking technologies - wired

Ethernet: modern Ethernet 100 Mbps or 1 Gbps

- Switched, full duplex and so no congestion
- No notion of bandwidth reservation
- Network is bounded by the packet processing duration, small packets require more per packet overhead (not amortized over the size of the packet)



Network technologies

- ATM allows for bandwidth reservation. ISPs love technology but for the most part, relegated to fringe networks.
- Wide area networks
 - Iptv topology courtesy Misha Rabinovich @CWRU



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