

Quality of Service and Multicasting in Computer Networks

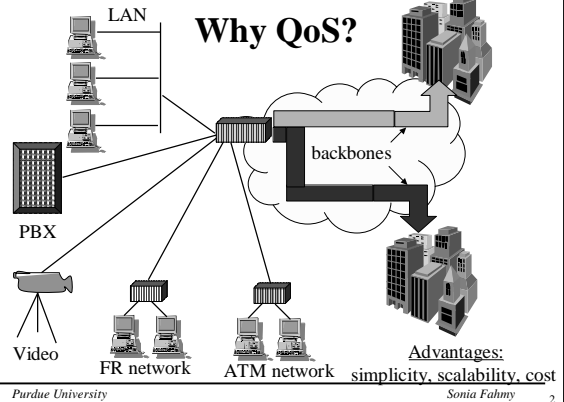
Sonia Fahmy

Department of Computer Sciences
Purdue University
fahmy@cs.purdue.edu

<http://www.cs.purdue.edu/homes/fahmy/>

(slides 4-6 and 10-20 used with permission of Raj Jain)

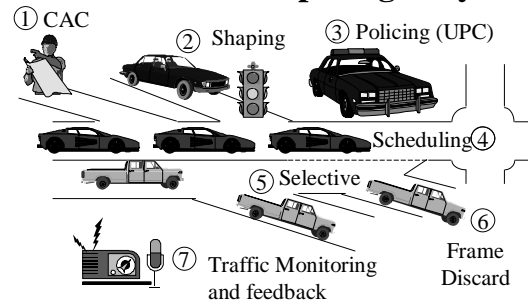
Why QoS?



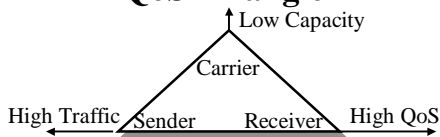
What is QoS?

- Predictable Quality: Throughput, Delay, Loss, Delay variation (jitter), Error rate
- Opposite of best effort (random quality)
- Mechanisms:
 - Capacity Planning: links, routers, switches
 - QoS based path determination, Route pinning
 - Admission and policy control, Signaling, Shaping, Policing (leaky and token buckets)
 - Classification, Queuing, Scheduling, Buffer management (drop thresholds and probabilities), Feedback and response to feedback

Traffic Management on the Information Superhighway



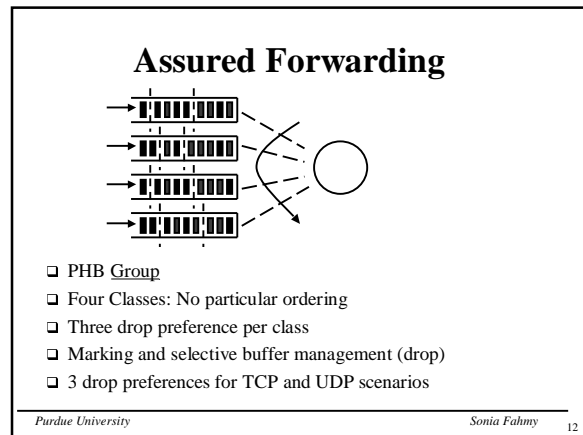
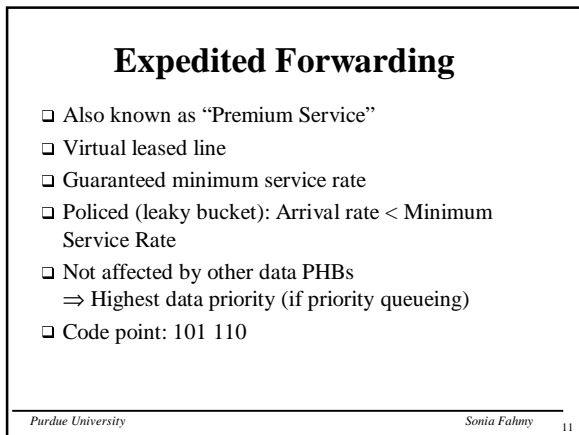
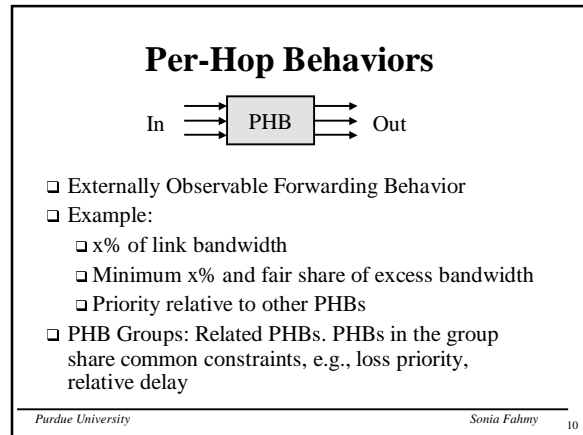
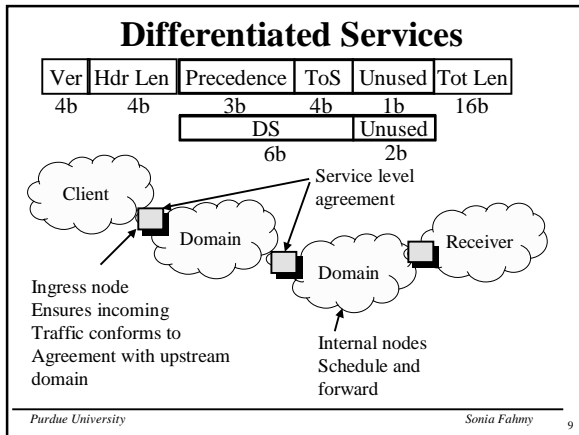
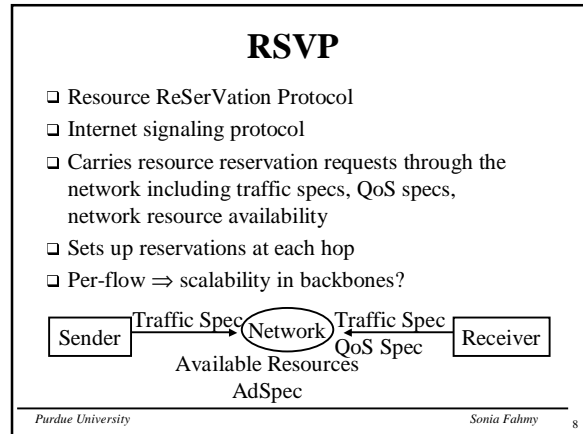
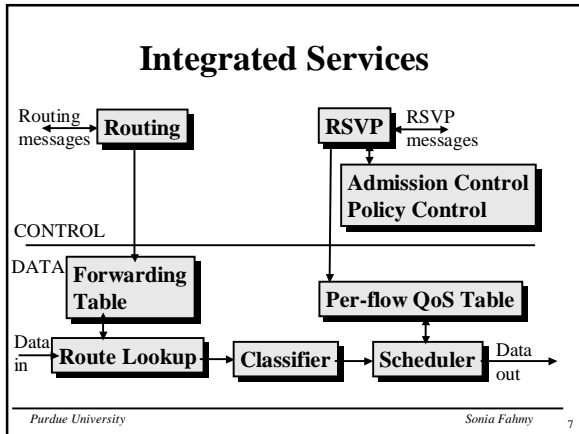
QoS Triangle



- Senders want to send traffic any time with high load, high burstiness
- Receivers expect low delay and high throughput
- Since links are expensive, providers want to minimize the infrastructure
- If one of the three gives in \Rightarrow no problem

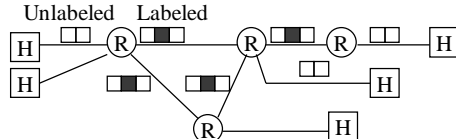
Integrated Services

- Best Effort Service
- Controlled-Load Service: Performance as good as in an unloaded datagram network. No quantitative assurances
- Guaranteed Service:
 - Firm bound on data throughput and delay.
 - Every element along the path must provide delay bound.
 - Is not always implementable, e.g., Shared Ethernet.



Multiprotocol Label Switching

- ❑ Label = Circuit number = VC Id
- ❑ Ingress router/host puts a label.
Exit router strips it off.
- ❑ Switches switch packets based on labels.
Do not need to look inside ⇒ Fast.



Purdue University

Sonia Fahmy

13

Traffic Engineering Objectives

- ❑ User's Performance Optimization
 - ⇒ Maximum throughput, Min delay, min loss, min delay variation
- ❑ Efficient resource allocation for the provider
 - ⇒ Efficient Utilization of all links
 - ⇒ Load Balancing on parallel paths
 - ⇒ Minimize buffer utilization
 - ❑ Current routing protocols (e.g., RIP and OSPF) find the shortest path (may be over-utilized).
- ❑ QoS Guarantee: Selecting paths that can meet QoS
- ❑ Enforce Service Level agreements
- ❑ Enforce policies: Constraint based routing \supseteq QoSR

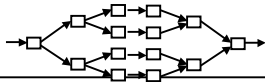
Purdue University

Sonia Fahmy

14

MPLS Mechanisms for TE

- ❑ Signaling, Admission Control, Routing
- ❑ Explicit routing of LSPs
- ❑ Constrained based routing of LSPs
Allows both Traffic constraints and Resource Constraints (Resource Attributes)
- ❑ Hierarchical division of the problem (Label Stacks)
- ❑ Traffic trunks allow aggregation and disaggregation (Shortest path routing allows only aggregation)

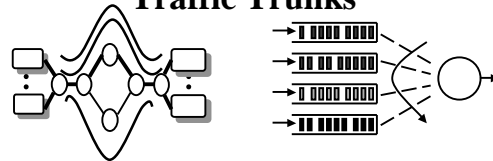


Purdue University

Sonia Fahmy

15

Traffic Trunks



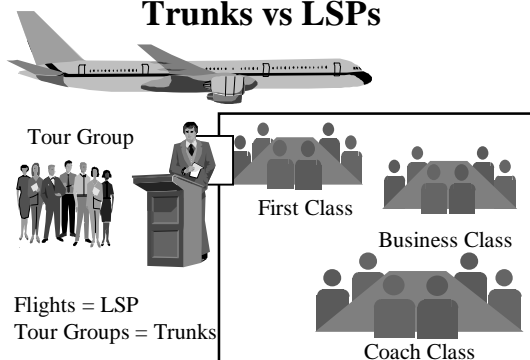
- ❑ Trunk: Aggregation of flows of same class on same LSP
- ❑ Trunks are routable
 - ⇒ LSP through which trunk passes can be changed
- ❑ Class ⇒ Queue, LSP ⇒ Next hop
Class can be coded in Exp or Label field. Assume Exp.

Purdue University

Sonia Fahmy

16

Trunks vs LSPs



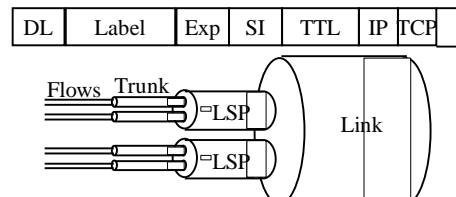
Purdue University

Sonia Fahmy

17

Flows, Trunks, LSPs, and Links

- ❑ Label Switched Path (LSP):
All packets with the same label
- ❑ Trunk: Same Label+Exp
- ❑ Flow: Same MPLS+IP+TCP headers



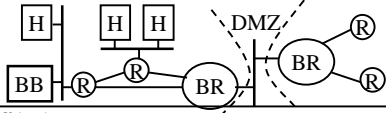
Purdue University

Sonia Fahmy

18

Bandwidth Broker

- ❑ Repository of policy database. Includes authentication
- ❑ Users request bandwidth from BB
- ❑ BB sends authorizations to leaf/border routers
Tells what to mark.
- ❑ Ideally, need to account for bandwidth usage along the path
- ❑ BB allocates only boundary or bottleneck

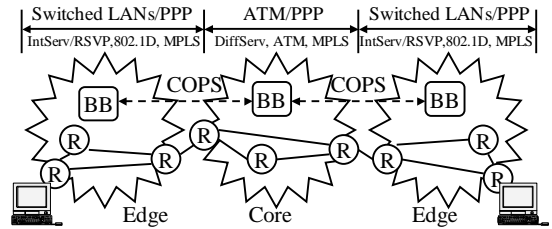


Purdue University

Sonia Fahmy 19

End-to-end View

- ❑ ATM/PPP backbone, Switched LANs/PPP in Stub
- ❑ IntServ/RSVP, 802.1D, MPLS in Stub networks
- ❑ DiffServ, ATM, MPLS in the core

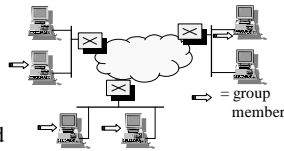


Purdue University

Sonia Fahmy 20

Why Multicast?

- ❑ Multipoint communication = exchange of information among multiple senders and multiple receivers (multicast group)
- ❑ Popular applications requiring multipoint support include:
 - ❑ conferencing, distance learning, software distribution, searching, server and database synchronization
- ❑ Complicated by variation in group size and dynamics and bandwidth requirements
- ❑ Using multicasting \Rightarrow traffic and management overhead not \propto number of participants



Purdue University

Sonia Fahmy 21

Multicast Requirements

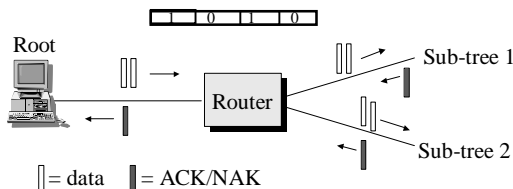
- ❑ Address formats (e.g., class D) and allocation
- ❑ Group membership management (e.g., IGMP)
- ❑ Routing (e.g., DVMRP)
- ❑ Forwarding and replication
- ❑ Flow control
- ❑ Congestion control

Purdue University

Sonia Fahmy 22

Reliable Multicast Transport

- ❑ **Consolidation** necessary to prevent implosion. Solutions: probabilistic suppression, scoping, NAKs or bitmaps, router assist, scheduling, hierarchy
- ❑ **Optimization** of repair traffic. Solutions: local retransmitters, subcasting, FEC

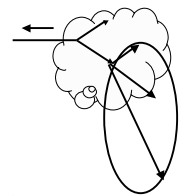


Purdue University

Sonia Fahmy 23

Multicast Challenges

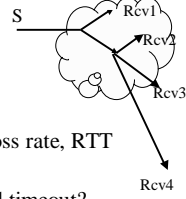
- ❑ **Goals:**
 - ❑ Fairness: TCP friendliness
 - ❑ High utilization of resources
 - ❑ Low loss and delay
 - ❑ Stability and fast transient response
 - ❑ Simplicity and generality (assume no layering)
- ❑ **Window-based** TCP-like multicast congestion control can under-estimate bandwidth share:
 - ❑ RTTmax wait before window update
 - ❑ Window set to minimum bandwidth-delay product
- ❑ \Rightarrow **Rate-based** preferred
- ❑ How to detect the congested sub-tree and its parameters?
 - ❑ Drop-to-zero problem: how to filter loss indications?



Purdue University

Sonia Fahmy 24

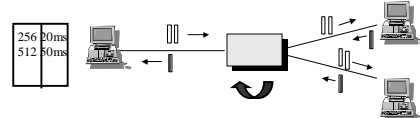
Multicast Challenges (cont)



- TCP friendliness:
 - TCP throughput is a function of loss rate, RTT and timeout values
- Where to estimate loss rate, RTT and timeout?
 - Receivers can compute rates and send them back to sender
 - Sender can estimate parameters more easily without extra traffic
- Should work with ACKs, NAKs, bitmaps, ECN, etc.

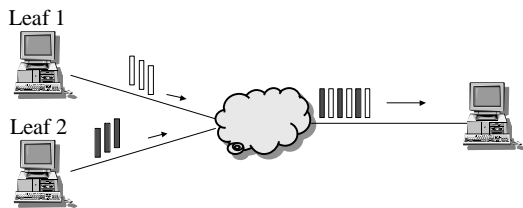
Approaches

- Sender-based parameter estimation for RTT and loss. Additive increase when no NAKs received during an RTT-dependent interval. Multiplicative decrease when NAK or ECN signaling a *new congestion epoch* received.
- Select an ACKer (pgmcc)
- Equation-based (Floyd and Handley)



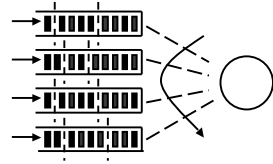
Approach (cont'd)

- Automatic split of heterogeneous groups depending on minimum and maximum rate thresholds
- Proactive FEC: increase FEC during high congestion
- Use-it-or-lose it for inactive senders and rate thresholds



Layered Multicast and Coordination

- RLM: receivers join appropriate layers
- RLC: sync. points
- Sender can use feedback to determine number and rates of layers (UCI/Rutgers)
- Router uses differential drop (AT&T)
- Coordinator determines TCP/UDP for each sender (Balakrishnan's CM (ecm IETF group), RACOON)



Thank You!



Questions?