# CS 356: Computer Network Architectures

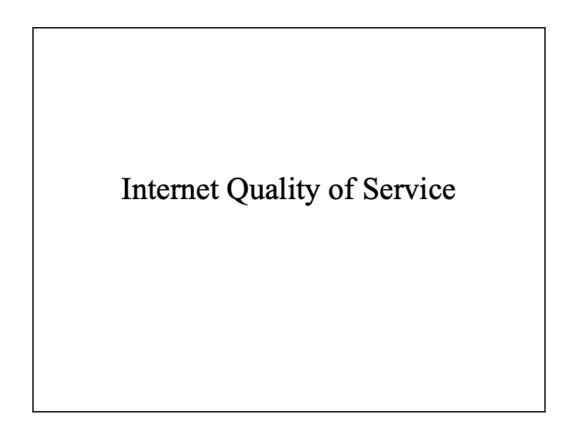
Lecture : Internet Quality of Service [PD] Chapter 6.5

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### Overview

- Network Resource Allocation
- Congestion Avoidance
- Why QoS?
  - Architectural considerations
- Approaches to QoS
  - Fine-grained: Integrated services
    - RSVP
  - Coarse-grained:
    - · Differentiated services
    - · Next lecture

Fine-grained: guarantees for each individual flow



### Motivation

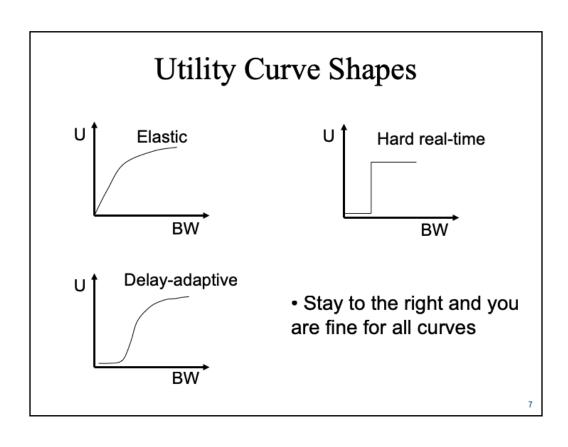
- Internet currently provides one single class of "best-effort" service
  - No assurance about delivery
- Many existing applications are *elastic* 
  - Tolerate delays and losses
  - Can adapt to congestion
- "Real-time" applications may be inelastic

# **Inelastic** Applications

- Continuous media applications
  - Lower and upper limit on acceptable performance
  - Below which video and audio are not intelligible
  - Internet telephones, teleconferencing with high delay (200 300ms) impair human interactions
- Hard real-time applications
  - Require hard limits on performance
  - E.g., industrial control applications
    - · Internet surgery

# Design question #1: Why a New Service Model?

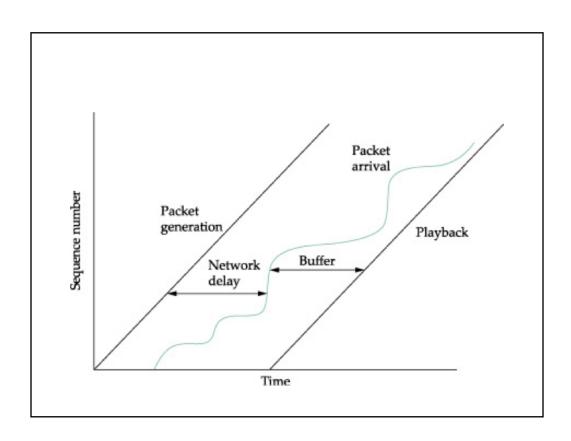
- What is the **basic objective** of network design?
  - Maximize total bandwidth? Minimize latency? Maximize ISP's revenues?
  - the designer's choice: Maximize social welfare: the total utility given to users (why not profit?)
- What does utility vs. bandwidth look like?
  - Must be non-decreasing function
  - Shape depends on application



# **Playback Applications**



- Sample signal → packetize → transmit → buffer → playback
  - Fits most multimedia applications
- Performance concern:
  - Jitter: variation in end-to-end delay
    - Delay = fixed + variable = (propagation + packetization) + queuing
- Solution:
  - Playback point delay introduced by buffer to hide network jitter



### Characteristics of Playback Applications

- In general lower delay is preferable
- Doesn't matter when packet arrives as long as it is before playback point
- Network guarantees (e.g., bound on jitter) would make it easier to set playback point
- Applications can tolerate some loss

### **Applications Variations**

- · Rigid and adaptive applications
  - Delay adaptive
    - · Rigid: set fixed playback point
    - · Adaptive: adapt playback point
      - E.g. Shortening silence for voice applications
  - Rate adaptive
- Loss tolerant and intolerant applications
- Four combinations

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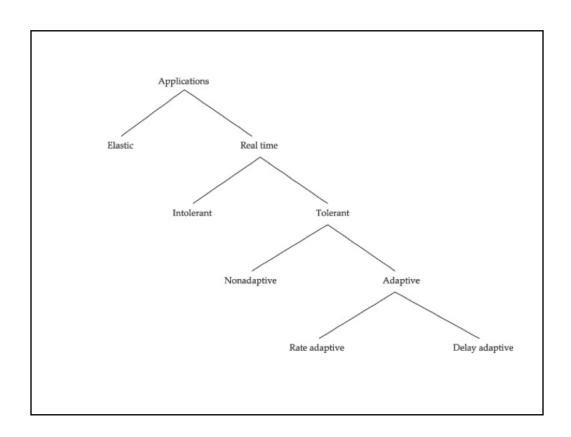
Move playback point around based on observed network delay

Gamble that network conditions will be the same as in the past

Are prepared to deal with errors in their estimate

Will have an earlier playback point than rigid applications

How to adapt:



## **Applications Variations**

#### Really only two classes of applications

- 1) Intolerant and rigid
- 2) Tolerant and adaptive

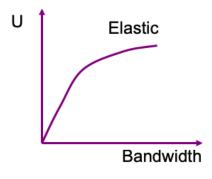
#### Other combinations make little sense

- 3) Intolerant and adaptive
  - Cannot adapt without interruption
- 4) Tolerant and rigid
  - Missed opportunity to improve delay

# Design question 2: How to maximize $V = \sum U(s_i)$

- Choice #1: add more pipes
- Choice #2: fix the bandwidth but offer different services
  - Q: can differentiated services improve V?

If all users' utility functions are elastic



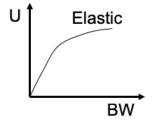
Does equal allocation of bandwidth maximize total utility?

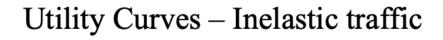
- $\sum s_i = B$
- Max  $\sum U(s_i)$

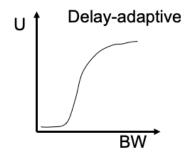
# Design question: is Admission Control needed?

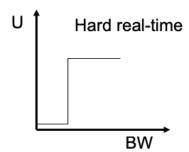
- If U(bandwidth) is concave
  - → elastic applications
    - Incremental utility is decreasing with increasing bandwidth
      - $U(x) = log(x^p)$
      - $V = nlog(B/n)^p = logB^p n^{1-p}$
    - Is always advantageous to have more flows with lower bandwidth
      - · No need of admission control;

This is why the Internet works! And fairness makes sense





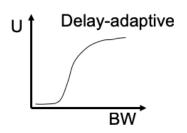




Does equal allocation of bandwidth maximize total utility?

## Is Admission Control needed?

- If U is convex → inelastic applications
  - U(number of flows) is no longer monotonically increasing
  - Need admission control to maximize total utility
- Admission control → deciding when the addition of new people would result in reduction of utility
  - Basically avoids overload



## **Incentives**

- Who should be given what service?
  - Users have incentives to cheat
  - Pricing seems to be a reasonable choice
  - But usage-based charging may not be well received by users

# Over provisioning

- Pros: simple
- Cons
  - Not cost effective
  - Bursty traffic leads to a high peak/average ratio
    - E.g., normal users versus leading edge users
  - It might be easier to block heavy users

## Comments

- End-to-end QoS has not happened
- Why?
- Can you think of any mechanism to make it happen?

# Approaches to QoS

- Fine-grained:
  - Integrated services
    - RSVP
- Coarse-grained:
  - Differentiated services

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Fine-grained: guarantees for each individual flow

## Components of Integrated Services

#### 1. Service classes

What does the network promise?

2. Service interface

How does the application describe what it wants?

3. Establishing the guarantee

How is the promise communicated to/from the network How is admission of new applications controlled?

4. Packet scheduling

How does the network meet promises?

## 1. Service classes

What kind of promises/services should network offer?



Depends on the **characteristics of the applications** that will use the network ....

#### Service classes

- Guaranteed service
  - For intolerant and rigid applications
  - Fixed guarantee, network meets commitment as long as clients send at match traffic agreement
- Controlled load service
  - For tolerant and adaptive applications
  - Emulate lightly loaded networks
- · Datagram/best effort service
  - Networks do not introduce loss or delay unnecessarily

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#### Two components

If conditions do not change, commit to current service

If conditions change, take steps to deliver consistent performance (help apps minimize playback delay)

Implicit assumption – network does not change much over time

## Components of Integrated Services

1. Type of commitment

What does the network promise?

2. Service interface

How does the application describe what it wants?

3. Establishing the guarantee

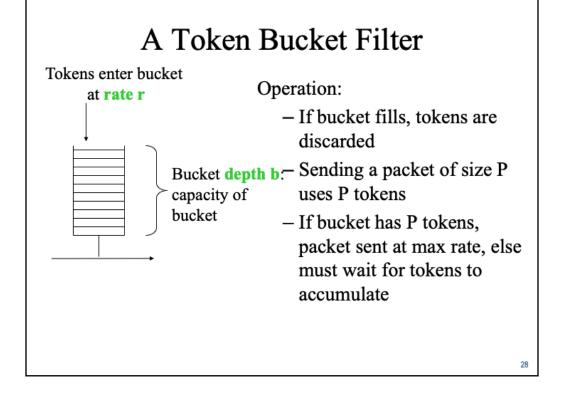
How is the promise communicated to/from the network How is admission of new applications controlled?

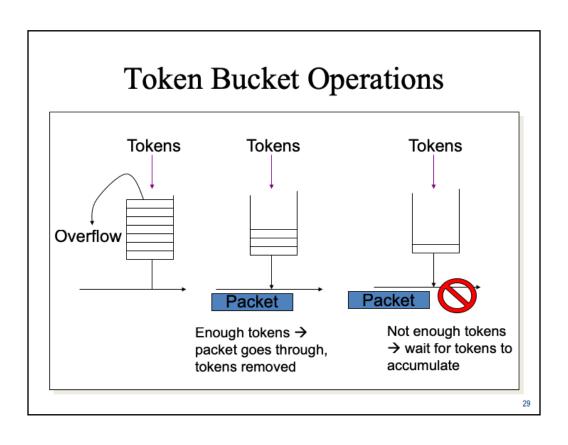
4. Packet scheduling

How does the network meet promises?

## Service interfaces

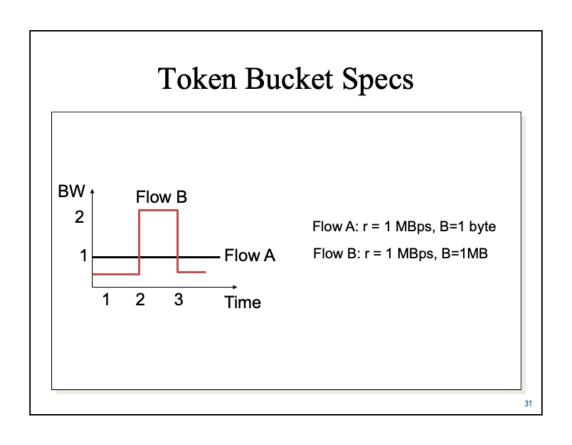
- Flowspecs
  - -TSpec: a flow's traffic characteristics
    - Difficult: bandwidth varies
  - RSpec: the service requested from the network
    - Service dependent
      - -E.g. controlled load





## **Token Bucket Characteristics**

- In the long run, rate is limited to r
- In the short run, a burst of size b can be sent
- Amount of traffic entering at interval T is bounded by:
  - Traffic = b + r\*T
- Information useful to admission algorithm



# **TSpec**

- TokenBucketRate
- TokenBucketSize
- PeakRate
- MinimumPolicedUnit
- MaximumPacketSize

# Service Interfaces: RSpec

- Guaranteed Traffic
  - TokenRate and DelayVariation
  - Or DelayVariation and Latency
- · Controlled load
  - Type of service

## Components of Integrated Services

1. Type of commitment

What does the network promise?

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How is the promise communicated to/from the network How is admission of new applications controlled?

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How does the network meet promises?

#### **RSVP** Goals

- Used on connectionless networks
  - Robust
  - Should not replicate routing functionality
  - Should co-exist with route changes
- Support for multicast
- Modular design should be generic "signaling" protocol
- Approaches
  - Receiver-oriented
  - Soft-state

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Different receivers have different capabilities and want different QOS

Changes in group membership should not be expensive

Reservations should be aggregate – I.e. each receiver in group should not have to reserve

Should be able to switch allocated resource to different senders

#### **RSVP Service Model**

- · Make reservations for simplex data streams
- Receiver decides whether to make reservation
- Control msgs in IP datagrams (proto #46)
- PATH/RESV sent periodically to refresh soft state

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# One pass:

Failed requests return error messages - receiver must try again No e2e ack for success

## **PATH Messages**

- PATH messages carry sender's Tspec
  - Token bucket parameters
- Routers note the direction PATH messages arrived and set up *reverse path* to sender
- Receivers send RESV messages that follow reverse path and setup reservations
- If reservation cannot be made, user gets an error

## **RESV Messages**

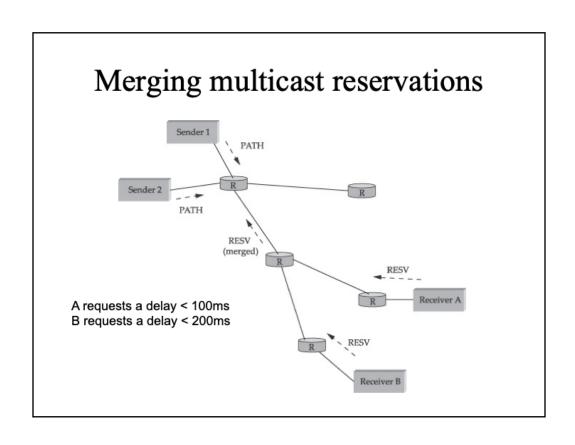
- Forwarded via reverse path of PATH
- A receiver sends RESV messages
  - TSpec from the sender
  - Rspec

## Admission control

- Router performs admission control and reserves resources
  - -If request rejected, send error message to receiver
  - -Guaranteed service: a yes/no based on available bandwidth
  - -Controlled load: heuristics
    - If delay has not exceeded the bound last time after admitting a similar flow, let it in

## Soft State to Adapt to Routing Changes

- Problems: Routing protocol makes routing changes
- Solution:
  - PATH and RESV messages sent periodically
  - Non-refreshed state times out automatically
- Ex: a link fails. How is a new reservation established?



## Components of Integrated Services

1. Type of commitment

What does the network promise?

2. Service interface

How does the application describe what it wants?

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How is the promise communicated to/from the network How is admission of new applications controlled?

4. Packet scheduling

How does the network meet promises?

# Packet classification and scheduling

- 1. Map a packet to a service class
  - (src addr, dst addr, proto, src port, dst port)
- 2. Use scheduling algorithms to provide the service
  - An implementation issue

## Scheduling for Guaranteed Traffic

- Use WFQ at the routers
  - Q: will DRR work?
- Each flow is assigned to its individual queue
- Parekh's bound for worst case queuing delay = b/r
  - -b = bucket depth
  - -r = rate of arrival

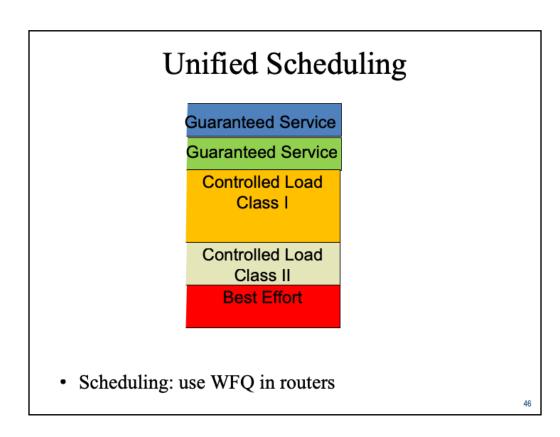
### **Controlled Load Service**

#### Goals:

- Isolation
  - Isolates well-behaved from misbehaving sources
- Sharing
  - Mixing of different sources in a way beneficial to all

#### **Possible Mechanisms:**

- WFQ
  - Aggregate multiple flows into one WFQ



## Each guaranteed flow gets its own queue

All controlled load service flows and best effort aggregates in single separate queue

Controlled load classes

Worst case delay for classes separated by order of magnitude

When high priority needs extra bandwidth – steals it from lower class

Best effort traffic acts as lowest priority class

## Scalability

- A lot of requests and state!
- ISPs feel it is not the right service model for them!
- Per-flow reservation/queue
  - OC-48 link 2.5Gbps
  - 64Kbps audio stream
  - $\rightarrow 39,000$  flows
  - Reservation and state needs to be stored in memory, and refreshed periodically
  - Classify, police, nd queue each flows

## Comments on RSVP

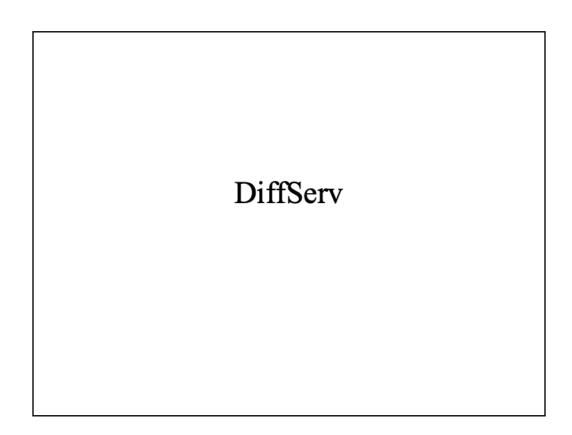
- Not widely deployed as a commercial service
- Used for other purposes
  - Setting up MPLS tunnels etc.

## Summary

- Why QOS?
  - Architectural considerations
- Approaches to QoS
  - Fine-grained: Integrated services
    - RSVP
  - Coarse-grained:
    - · Differentiated services
- Next lecture:
  - DiffServ
  - Net Neutrality

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Fine-grained: guarantees for each individual flow



### Motivation of DiffServ

- Analogy:
  - Airline service, first class, coach, various restrictions on coach as a function of payment
- · Economics and assurances
  - Pay more, and get better service
  - Best-effort expected to make up bulk of traffic,
  - Revenue from first class important to economic base
  - Not motivated by real-time or maximizing social welfare

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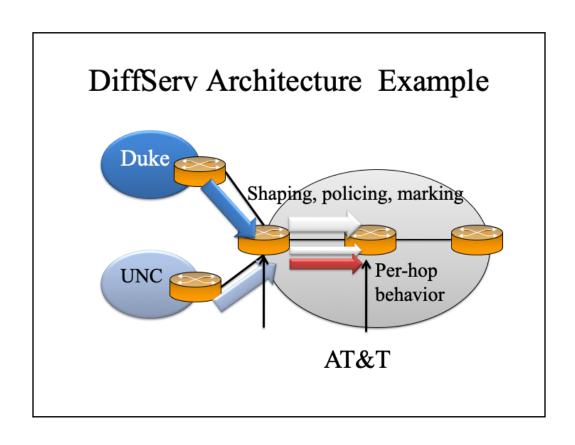
(will pay for more plentiful bandwidth overall)

#### **Basic Architecture**

- Agreements/service provided within a domain
  - Service Level Agreement (SLA) with ISP
- Edge routers do traffic conditioning
  - Shaping, Policing, and Marking
- Core routers
  - Process packets based on packet marking and defined per hop behavior (PHB)
- More scalable than IntServ
  - No per flow state or signaling

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Edge routers: Perform per aggregate shaping and policing
Mark packets with a small number of bits; each bit encoding represents
a class or subclass



## Per-hop Behaviors (PHBs)

- Define behavior of individual routers rather than endto-end services; there may be many more services than behaviors
  - No end-to-end guarantee
- Multiple behaviors need more than one bit in the header
- Six bits from IP TOS field are taken for Diffserv code points (DSCP)

## Per-hop Behaviors (PHBs)

- Two PHBs defined so far
- Expedited forwarding aka premium service (type P)
  - Possible service: providing a virtual wire
- Assured forwarding (type A)
  - Possible service: strong assurance for traffic within profile and allow source to exceed profile

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# Admitted based on peak rate Unused premium goes to best effort

Based on expected capacity usage profiles

Traffic unlikely to be dropped if user maintains profile

Out-of-profile traffic marked

## **Expedited Forwarding PHB**

• Goal: EF packets are forwarded with minimal delay and loss

#### • Mechanisms:

- User sends within profile and network commits to delivery with requested profile
- Rate limiting of EF packets at edges only, using token bucket to shape transmission
- Priority or Weighted Fair Queuing

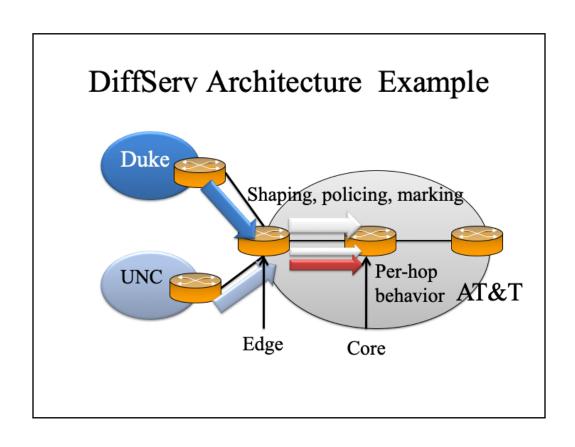
## **Assured Forwarding PHB**

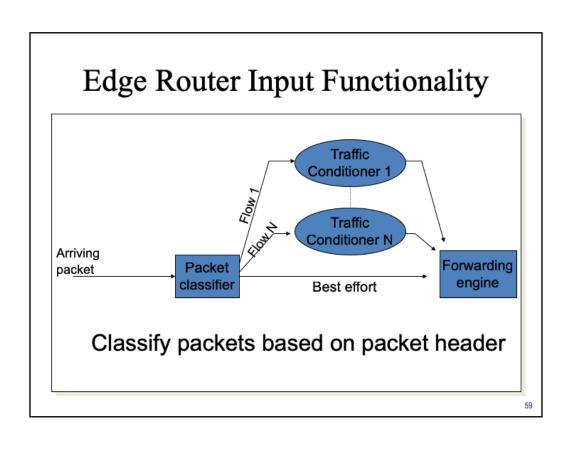
- Goal: good services for in-profile traffic
- Mechanisms:
  - User and network agree to some traffic profile
    - · How to define profiles is an open/policy issue
  - Edges mark packets up to allowed rate as "inprofile" or low drop precedence
  - Other packets are marked with one of two higher drop precedence values
  - Random Early Detection in/out queues

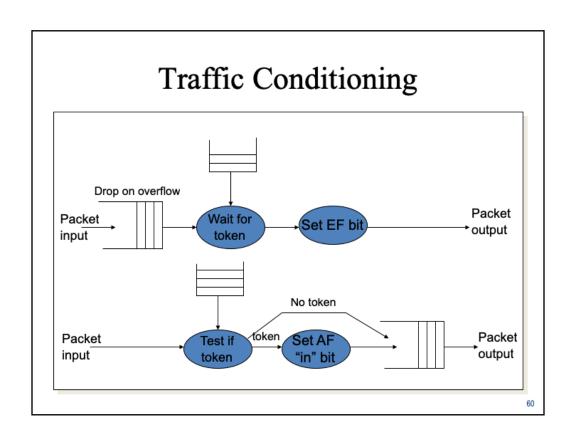
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A congested DS node tries to protect packets with a lower drop precedence value from being lost by preferably discarding packets with a higher drop precedence value

Implemented using RED with In/Out bit

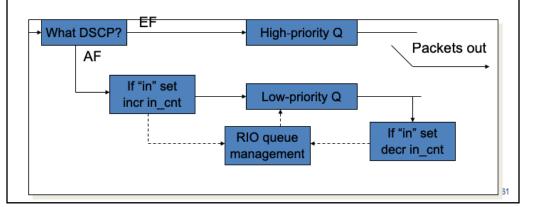






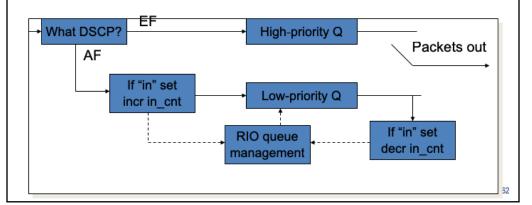
## **Router Output Processing**

- Two queues: EF packets on higher priority queue
- Lower priority queue implements RED "In or Out" scheme (RIO)



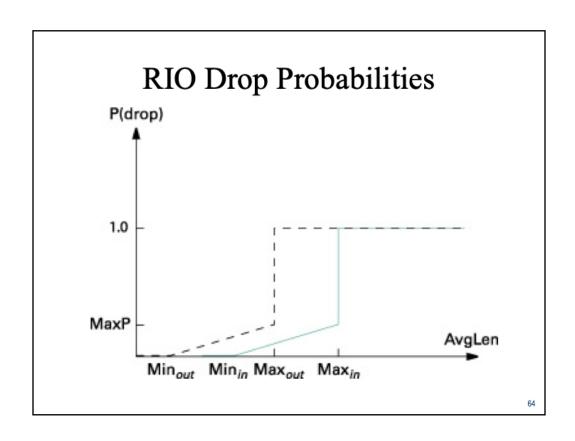
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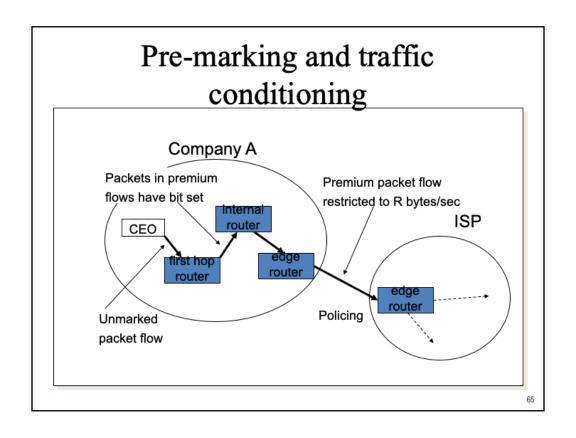
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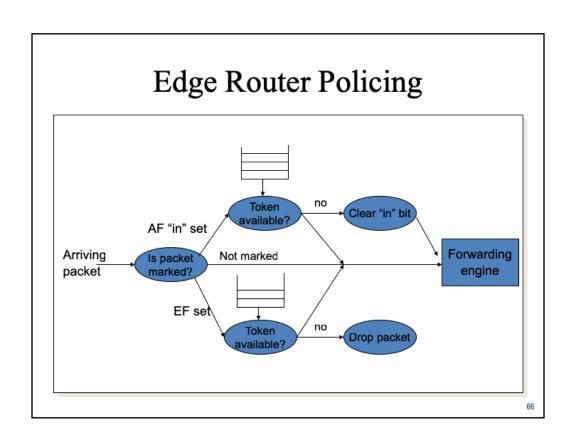


## Red with In or Out (RIO)

- Similar to RED, but with two separate probability curves
- Has two classes, "In" and "Out" (of profile)
- "Out" class has lower Min<sub>thresh</sub>, so packets are dropped from this class first
  - Based on queue length of all packets
- As avg queue length increases, "in" packets are also dropped
  - Based on queue length of only "in" packets







## Remarks on QoS

- "Dead" at the Internet scale
- Areas of success
  - Enterprise networks
  - Residential uplinks
  - Datacenter networks

## Conclusion

- Multicast
  - Service model
  - Sample routing protocols
- QoS
  - Why do we need it?
  - Integrated Services
  - Differentiated Services
    - · Motivated by business models