Overview

- Why QoS?
- Integrated services
- Internet video
- Differentiated services

Motivation

- Internet currently provides one single class of "best-effort" service
  - No assurances about delivery
- Existing applications are elastic
  - Tolerate delays and losses
  - Can adapt to congestion
- Future "real-time" applications may be inelastic

QoS

- IntServ
- DiffServ
- Assigned reading
  - [She95] Fundamental Design Issues for the Future Internet
- Optional
  - [CSZ92] Supporting Real-Time Applications in an Integrated Services Packet Network: Architecture and Mechanisms
  - [CF98] Explicit Allocation of Best-Effort Packet Delivery Service
### Inelastic Applications

- **Continuous media applications**
  - Lower and upper limit on acceptable performance.
  - BW below which video and audio are not intelligible
  - Internet telephones, teleconferencing with high delay (200 - 300ms) impair human interaction

- **Hard real-time applications**
  - Require hard limits on performance
  - E.g. control applications

### Why a New Service Model?

- What is the **basic objective** of network design?
  - Maximize total bandwidth? Minimize latency?
  - **Maximize user satisfaction** – the total utility given to users

- What does utility vs. bandwidth look like?
  - Must be non-decreasing function
  - Shape depends on application

### Utility Curve Shapes

- **Elastic**
- **Hard real-time**
- **Delay-adaptive**

Stay to the right and you are fine for all curves

### Utility curve – Elastic traffic

Does equal allocation of bandwidth maximize total utility?
Admission Control

- If $U$ (bandwidth) is concave → elastic applications
  - Incremental utility is decreasing with increasing bandwidth
  - Is always advantageous to have more flows with lower bandwidth
    - No need of admission control; This is why the Internet works!

Utility Curves – Inelastic traffic

- 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

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Components of Integrated Services

1. **Type of commitment**
   What does the network promise?
2. Packet scheduling
   How does the network meet promises?
3. Service interface
   How does the application describe what it wants?
4. Establishing the guarantee
   How is the promise communicated to/from the network
   How is admission of new applications controlled?

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 & adaptive applications
  - Rigid – set fixed playback point
  - Adaptive – adapt playback point
    - 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

- Tolerant & intolerant applications
  - Tolerance to brief interruptions in service

- 4 combinations

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

So what service classes should the network offer?

Components of Integrated Services

1. Type of commitment
   - What does the network promise?

2. Packet scheduling
   - How does the network meet promises?

3. Service interface
   - How does the application describe what it wants?

4. Establishing the guarantee
   - How is the promise communicated to/from the network
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Type of Commitments

- **Guaranteed** service
  - For intolerant and rigid applications
  - Fixed guarantee, network meets commitment as long as clients send at match traffic agreement

- **Predicted** service
  - For tolerant and adaptive applications
  - 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

- **Datagram/best effort service**
Scheduling for Guaranteed Traffic

- Use **token bucket filter** to characterize traffic
  - Described by rate $r$ and bucket depth $b$
- Use **WFQ** at the routers
- Parekh’s bound for worst case queuing delay = $b/r$

Token Bucket Filter

Operation:
- If bucket fills, tokens are discarded
- Sending a packet of size $P$ uses $P$ tokens
- If bucket has $P$ tokens, packet sent at max rate, else must wait for tokens to accumulate

Token Bucket Operation

- Tokens enter bucket at rate $r$
- Bucket depth $b$: capacity of bucket

Token Bucket Characteristics

- On the long run, rate is limited to $r$
- On the short run, a burst of size $b$ can be sent
- Amount of traffic entering at interval $T$ is bounded by:
  - Traffic = $b + rT$
- Information useful to admission algorithm
Token Bucket Specs

<table>
<thead>
<tr>
<th>BW</th>
<th>Flow A</th>
<th>Flow B</th>
</tr>
</thead>
<tbody>
<tr>
<td>1</td>
<td>1 2 3</td>
<td>2</td>
</tr>
</tbody>
</table>

Flow A: r = 1 MBps, B=1 byte
Flow B: r = 1 MBps, B=1MB

Predicted Service

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

Mechanisms:
- WFQ
  - Great isolation but no sharing
- FIFO
  - Great sharing but no isolation

Predicted Service

- FIFO jitter increases with the number of hops
  - Use opportunity for sharing across hops
- FIFO+
  - At each hop: measure average delay for class at that router
  - For each packet: compute difference of average delay and delay of that packet in queue
  - Add/subtract difference in packet header
  - Packet inserted into queues expected arrival time instead of actual
    - More complex queue management!
  - Slightly decreases mean delay and significantly decreases jitter

Unified Scheduling

- Assume 3 types of traffic: guaranteed, predictive, best-effort
- Scheduling: use WFQ in routers
- Each guaranteed flow gets its own queue
- All predicted service flows and best effort aggregates in single separate queue
  - Predictive traffic classes
    - Multiple FIFO+ queues
    - 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
Service Interfaces

- Guaranteed Traffic
  - Host specifies rate to network
  - Why not bucket size b?
    - If delay not good, ask for higher rate
- Predicted Traffic
  - Specifies (r, b) token bucket parameters
  - Specifies delay D and loss rate L
  - Network assigns priority class
  - Policing at edges to drop or tag packets
    - Needed to provide isolation – why is this not done for guaranteed traffic?
      - WFQ provides this for guaranteed traffic

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Internet Video Today

- Client-server streaming
  - Skype video conferencing
  - Hulu
- DVD transfer
  - BitTorrent → P2P lecture
- Synchronized video (IPTV)
  - Overlay multicast → multicast lecture

Client-Server Streaming: Adaptation Quality to Link

Long Time Scale

Short Time Scale

Content Negotiation

Server Selection

Adaptive Media

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Problems Adapting to Network State

- TCP hides network state
- New applications may not use TCP
  - Often do not adapt to congestion

Need system that helps applications learn and adapt to congestion

Congestion Manager Architecture

Transmission API

- Buffered send
  - cm_send(data, length)
- Request/callback-based send
  - cm_request()
  - cmapp_send()
  - cm_notify(nsent)

Transmission API (cont.)

- Request API: asynchronous sources
  ```
  wait for (some_events) {
    get_data();
    send();
  }
  ```
- Synchronous sources
  ```
  do_every_t_ms {
    get_data();
    send();
  }
  ```
- Solution: cmapp_update(rate, srtt) callback
Feedback about Network State

• Monitoring successes and losses
  • Application hints
  • Probing system

• Notification API (application hints)
  • Application calls cm_update(nsent, nrecd, congestion indicator, rtt)

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DiffServ

• Analogy:
  • Airline service, first class, coach, various restrictions on coach as a function of payment
• Best-effort expected to make up bulk of traffic, but revenue from first class important to economic base (will pay for more plentiful bandwidth overall)
• Not motivated by real-time! Motivated by economics and assurances

Basic Architecture

• Agreements/service provided within a domain
  • Service Level Agreement (SLA) with ISP
• Edge routers do traffic conditioning
  • Perform per aggregate shaping and policing
  • Mark packets with a small number of bits; each bit encoding represents a class or subclass
• Core routers
  • Process packets based on packet marking and defined per hop behavior
• More scalable than IntServ
  • No per flow state or signaling
Per-hop Behaviors (PHBs)

- Define behavior of individual routers rather than end-to-end services – there may be many more services than behaviors
- Multiple behaviors – need more than one bit in the header
- Six bits from IP TOS field are taken for Diffserv code points (DSCP)

Two PHBs defined so far
- Expedited forwarding aka premium service (type P)
  - Possible service: providing a virtual wire
  - Admitted based on peak rate
  - Unused premium goes to best effort
- Assured forwarding (type A)
  - Possible service: strong assurance for traffic within profile & allow source to exceed profile
  - Based on expected capacity usage profiles
  - Traffic unlikely to be dropped if user maintains profile
  - Out-of-profile traffic marked

Expedited Forwarding PHB

- User sends within profile & network commits to delivery with requested profile
  - Signaling, admission control may get more elaborate in future
- Rate limiting of EF packets at edges only, using token bucket to shape transmission
- Simple forwarding: classify packet in one of two queues, use priority
  - EF packets are forwarded with minimal delay and loss (up to the capacity of the router)

Expedited Forwarding Traffic Flow

Company A

Packets in premium flows have bit set

ISP

Unmarked packet flow

Host

First hop router

Internal router

Edge router

Premium packet flow restricted to R bytes/sec
**Assured Forwarding PHB**
- User and network agree to some traffic profile
  - Edges mark packets up to allowed rate as “in-profile” or low drop precedence
  - Other packets are marked with one of 2 higher drop precedence values
- 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

**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\(_{\text{thresh}}\), 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

**RIO Drop Probabilities**

<table>
<thead>
<tr>
<th></th>
<th>P (drop in)</th>
<th>P (drop out)</th>
</tr>
</thead>
<tbody>
<tr>
<td>min_in</td>
<td>P max_in</td>
<td>max_out</td>
</tr>
<tr>
<td>max_in</td>
<td>avg_in</td>
<td>avg_total</td>
</tr>
<tr>
<td>avg_in</td>
<td></td>
<td></td>
</tr>
</tbody>
</table>

**Edge Router Input Functionality**

- Arriving packet
- Packet classifier
- Best effort
- Forwarding engine

classify packets based on packet header
Traffic Conditioning

Router Output Processing
- 2 queues: EF packets on higher priority queue
- Lower priority queue implements RED “In or Out” scheme (RIO)

Edge Router Policing

Comparison

<table>
<thead>
<tr>
<th></th>
<th>Best-Effort</th>
<th>DiffServ</th>
<th>IntServ</th>
</tr>
</thead>
<tbody>
<tr>
<td>Service</td>
<td>• Connectivity</td>
<td>• Per aggregation isolation</td>
<td>• Per flow isolation</td>
</tr>
<tr>
<td></td>
<td>• No isolation</td>
<td>• Per aggregation guarantee</td>
<td>• Per flow guarantee</td>
</tr>
<tr>
<td>Service Scope</td>
<td>• End-to-end</td>
<td>• Domain</td>
<td>• End-to-end</td>
</tr>
<tr>
<td>Complexity</td>
<td>• No set-up</td>
<td>• Long term setup</td>
<td>• Per flow setup</td>
</tr>
<tr>
<td>Scalability</td>
<td>• Highly scalable (nodes maintain only routing state)</td>
<td>• Scalable (edge routers maintain per aggregate state; core routers per class state)</td>
<td>• Not scalable (each router maintains per flow state)</td>
</tr>
</tbody>
</table>
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Role of RSVP

• Rides on top of unicast/multicast routing protocols
• Carries resource requests all the way through the network
• At each hop consults admission control and sets up reservation. Informs requester if failure

RSVP Goals

• Used on connectionless networks
  • Should not replicate routing functionality
  • Should co-exist with route changes
• Support for multicast
  • 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
• Modular design – should be generic “signaling” protocol
• Result
  • Receiver-oriented
  • Soft-state
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
- 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
- Queuing delay and bandwidth requirements
- Source traffic characteristics (from PATH)
- Filter specification
  - Which transmissions can use the reserved resources
- Router performs admission control and reserves resources
  - If request rejected, send error message

PATH and RESV Messages
Routing Changes

- Routing protocol makes routing changes
- In absence of route or membership changes, periodic PATH and RESV msgs refresh established reservation state
- When change, new PATH msgs follow new path, new RESV msgs set reservation
- Non-refreshed state times out automatically

Why a New Service Model?

- Given the shape of different utility curves – clearly equal allocation of bandwidth does not maximize total utility
- In fact, desirable rate for some flow may be 0.

Admission Control

- Caveats
  - Admission control can only turn away new requests → sometimes it may be have been better to terminate an existing flow
  - U(0) ≠ 0 → users tend to be very unhappy with no service – perhaps U should be discontinuous here
- Alternative → overprovision the network
  - Problem: high variability in usage patterns
  - "Leading-edge" users make it costly to overprovision
  - Having admission control seems to be a better alternative
Other QOS principles

1. Admission Control
2. Marking of packets is needed to distinguish between different classes.
3. Protection (isolation) for one class from another.
4. While providing isolation, it is desirable to use resources as efficiently as possible → sharing.

How to Choose Service – Implicit

Network could examine packets and implicitly determine service class

- No changes to end hosts/applications
- Fixed set of applications supported at any time
- Can’t support applications in different uses/modes easily
- Violates layering/modularity

How to Choose Service – Explicit

Applications could explicitly request service level

- Why would an application request lower service?
  - Pricing
  - Informal social conventions
  - Problem exists in best-effort as well → congestion control

- Applications must know network service choices
  - Difficult to change over time
  - All parts of network must support this → places greater burden on portability of IP

Parekh Bound on Delay Across Net

\[ D_i = \left( \frac{\text{bucket size}}{\text{weighted rate allocated}} \right) + \left[ \frac{(\text{nhops} - 1) \times \text{MaxPacketLen}}{\text{weighted rate allocation}} \right] + \sum_{m=1}^{\text{hop}_i} \left( \frac{\text{max packet length}}{\text{outbound bw at hop}} \right) \]

- 1st term: delay when running at full speed
- 2nd term: packetization effects
- 3rd term: added delay due to packet approx of FQ (goes away as data rate increases)
IETF Internet Service Classes

- Guaranteed service
  - Firm bounds on e2e delays and bandwidth
- Controlled load
  - “A QoS closely approximating the QoS that same flow would receive from an unloaded network element, but uses capacity (admission) control to assure that this service is received even when the network element is overloaded”
- Best effort

Reservation Protocol: RSVP

- Upper layer protocols and applications
  - IP service interface
    - IP
      - ICMP
      - IGMP
      - RSVP
  - Link layer service interface
  - Link layer modules

Basic Message Types

- PATH message
- RESV message
- CONFIRMATION message
  - Generated only upon request
  - Unicast to receiver when RESV reaches node with established state
- TEARDOWN message
- ERROR message (if PATH or RESV fails)

Packet Classifying and Scheduling

- Each arriving packet must be:
  - Classified: associated with the application reservation
    - Fields: source + destination address, protocol number, source + destination port
  - Scheduled: managed in the queue so that it receives the requested service
    - Implementation not specified in the service model, left up to the implementation
RSVP and Multicast

- Reservations from multiple receivers for a single sender are merged together at branching points
- Reservations for multiple senders may not be added up:
  - Audio conference, not many talk at the same time
  - Only subset of speakers (filters)
  - Mixers and translators

Reservation Styles

- How filters are used
- Three styles
  - Wildcard/No filter – does not specify a particular sender for group
  - Fixed filter – sender explicitly specified for a reservation
  - Dynamic filter – valid senders may be changed over time
- Receiver chooses but sender can force no-filter by setting F-Flag

Changing Reservation

- Receiver-oriented approach and soft state make it easy to modify reservation
- Modification sent with periodic refresh

Possible Token Bucket Uses

- Shaping, policing, marking
  - Delay pkts from entering net (shaping)
  - Drop pkts that arrive without tokens (policing)
  - Let all pkts pass through, mark ones without tokens
    - Network drops pkts without tokens in time of congestion
Guarantee Proven by Parekh

- Given:
  - Flow $i$ shaped with token bucket and leaky bucket rate control (depth $b$ and rate $r$)
  - Network nodes do WFQ
  - Cumulative queuing delay $D_i$ suffered by flow $i$ has upper bound
    - $D_i < b/r$, (where $r$ may be much larger than average rate)
    - Assumes that $\bar{X}r < \text{link speed at any router}$
    - All sources limiting themselves to $r$ will result in no network queuing

Establishing the guarantee

- Admission control
  - Don’t give all bandwidth to real-time traffic
    - 90% real-time, 10% best effort
  - Very much dependent on how large fluctuations in network traffic and delay are
    - Should measure this dynamically instead of having built-in assumptions