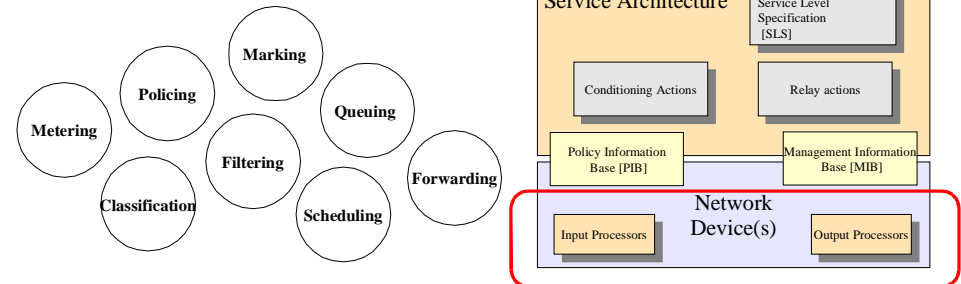


S–38.180 Palvelunlaatu Internetissä S–38.180 Quality of Service in Internet

Luento 3: Mekanismit – osa 1 Lecture 3: Mechanisms – part 1

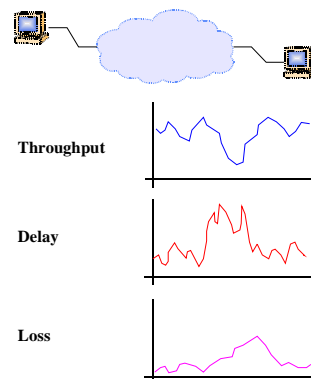
Today's Topic

- This lecture is about functional mechanisms which can be found from the input/output processors of network devices



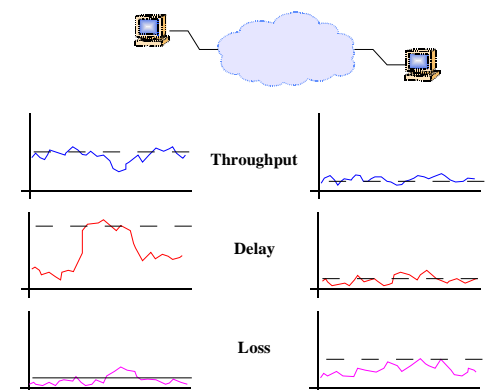
Internet Service

- Internet service is packet delivery service – pretty much like common snailmail service
 - Datagrams are delivered as individual items and they experience service which varies as a function of time



Internet Service

- By adding QoS, we are aiming to provide differentiation to this situation
- Differentiation can be based on different criterion
 - Usage
 - Money
 - Status





Terminology

- **Connection:** is dynamically formed reservation of network resources for a period of time.
 - Connection requires a **state** to be formed inside the network
 - State is a filter defining packets which belong into particular connection and required reservation attributes
- **Flow:** is formed from arbitrary packets which fall within predefined filter and temporal behavior.
 - Packets from one source to same destination arrive to investigation point with interarrival time less than t seconds.



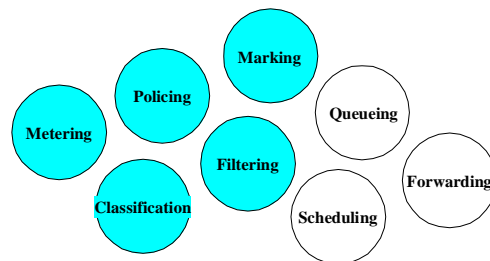
Terminology

- **Aggregate:** is a group of flows which have same forwarding characteristics and share link resources.
- **Class:** is a group of connections which share same forwarding characteristics.



Input processor

- Input processor of Internet router consists several mechanisms
 - Filtering
 - Classification
 - Metering
 - Policing
 - Marking
 - Shaping

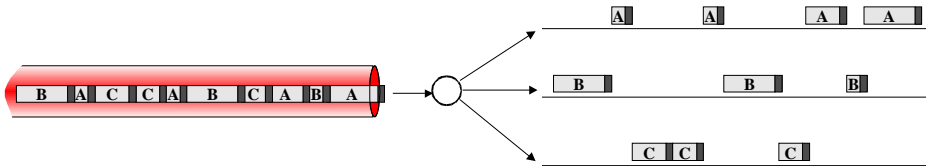


Classification

- Individual **connections** can be recognized by looking sufficient number of protocol fields.
- This is used in Integrated Services architecture.
- IntServ uses reservation protocol for informing the network about fields which should be examined.
- If per connection accuracy is not needed or can not be feasibly implemented is **class** based operation the answer.
- This is used in Differentiated Services architecture.
- Class is based on static filters covering broad range of different connections i.e. aggregating connections to one logical unit

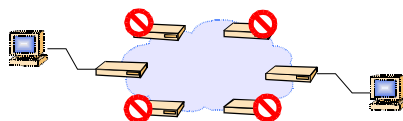
Classification

- **Classification** is process where packets in the packets stream are separated into *n* logically separate packet streams.
- These streams are then treated as separate entities for which different actions are performed
- Separation is based on **filters** which match packet content to the filtering rules.



Service Level Management

- QoS based networks need careful management
 - How to provision the network so that there will not be unnecessary queuing or packet loss
 - How to control the amount of traffic that gets into the network
- Network level
- Customer level / connection level
- Packet level



Filtering

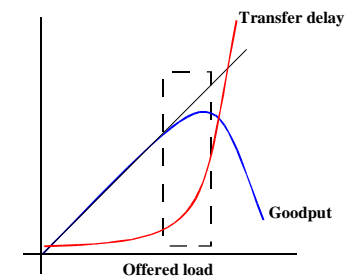
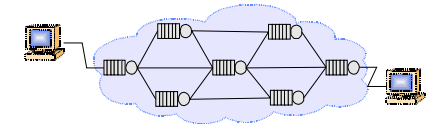
- Commonly filters are based on IP packet / transport header information
 - IP addresses
 - Protocol information
 - DSCP-field
 - Port information
 - Length information

Version	IHL	ToS/DSCP	Flags	Length
Identification		Offset		
TTL	Protocol	Checksum		
Source Address				
Destination Address				
Options			Padding	
Source Port			Destination Port	

- Generally **any fixed block of bits** can be used as a filter

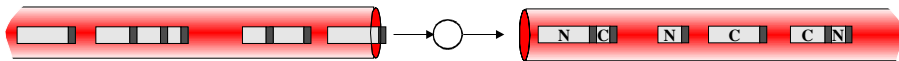
Service Level Management

- Overall objective is to offer QoS and/or maximize network throughput
- This requires
 - Limiting user traffic to the level that individual links operate on optimal fashion
 - Individual links can not be fully utilized
 - Unequal capacities
 - Uncertainty of paths
 - Uncertainty of demands



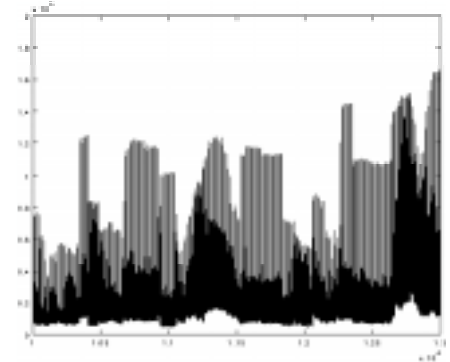
Rate Control

- Task is to decide which user packets should be **delivered** into the network and on what priority (**mark**)
 - They do not violate QoS management principles within the network by overloading the network
- Rate control operates in three levels
 - **Measures** the traffic
 - **Compares** the measured information to information in user / network policy
 - **Executes** policy based on comparison results
 - Marking
 - Dropping
 - Shaping



Rate Control

- User traffic process is largely dependent on application which is used.
 - Some applications produce constant traffic stream
 - Fixed size packets
 - Constant interarrival times
 - Other may produce bursts of packets
 - Variable size packets
 - Variable interarrival times



Rate Control

- Objectives:
 - Simple
 - Easy algorithm
 - Few parameters
 - Accurate
 - Actions are correct
 - Actions are transparent
 - Actions are immediate
 - Predictable
 - Action are consistent from time to time
- Requires:
 - Parametrization of user traffic
 - Either flow level
 - Or Aggregate level
 - This is bound to SLA made with the ISP

Metering

- Packet stream is measured to find out some of the following parameters:
 - Peak rate – maximum rate on which user is sending
 - Sustained rate – average rate on which user is sending
 - Burst size – maximum burst size which user sending on either with peak or average rate
- Actual measurement of information may be based on
 - Continuous time measurement
 - Discrete event analysis
 - Window based analysis

Token Bucket

- Produces information whether arrival rate is more or less than the threshold
- Algorithm is based on
 - Number of tokens in token bucket (in bytes)
 - Arrival time (T_{Now} , $T_{Last\ Arrival}$)
- Two limiting parameters
 - Bucket size (S)
 - Token rate (R) * token size

Initial condition:

Number of Tokens = S

Upon each arrival:

Increment = $TokenSize \cdot R \cdot (T_{Now} - T_{Last\ Arrival})$

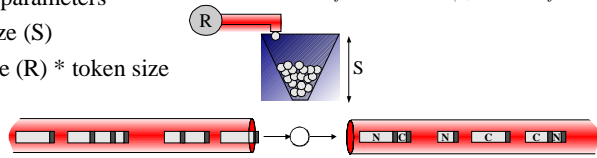
Decrement = PacketLength

Conformance = Number of Tokens + Increment - Decrement

if Conformance ≥ 0

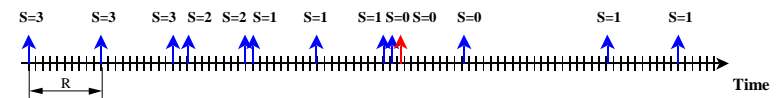
then Number of Tokens = $\min(S, Conformance)$

else Number of Tokens = $\min(S, Number\ of\ Tokens + Increment)$



Token Bucket

- In ideal situation
 - Packets arrive with intervals of token generation rate (R)
 - Packets are size of token
 - Variation of arrivals is compensated with bucket size (S)
 - Allows bursting
- Example:
 - R=10
 - S=3



Packet per packet EWMA meter

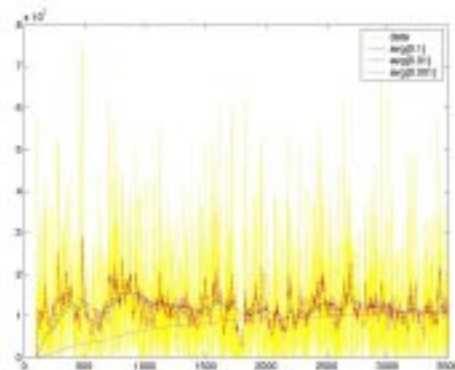
- Measures packet stream by using exponentially weighted moving average filter.
 - Tunable by parameter
 - Memory (ϵ)

Initial condition:

$avg(0) = 0$

After every packet arrival

$$avg(n+1) = (1-\epsilon) \cdot avg(n) + \epsilon \cdot \frac{PacketLength}{t_{n+1} - t_n}$$



Windowed EWMA meter

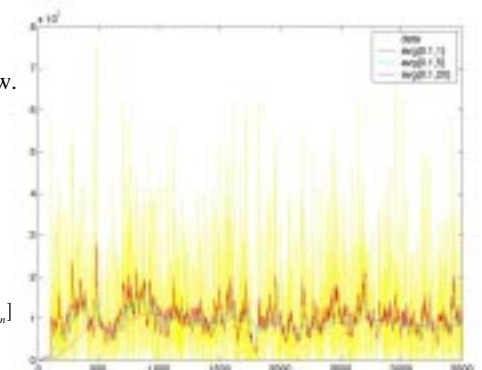
- Measures packet stream by using exponentially weighted moving average filter with sampling window.
 - Tunable by parameters
 - Memory (ϵ)
 - Sampling interval (ΔT)

Initial condition:

$avg(0) = 0$

After every ΔT time units

$$avg(t_{n+1}) = (1-\epsilon) \cdot avg(t_n) + \epsilon \cdot \text{bytes during } [t_{n+1}, t_n]$$



Time Sliding Window Meter

- TSW is memory based, windowed average rate estimator
- Tunable by parameter
 - Window length

Initial condition:

$$avg(0) = 0$$

$$Win_{length} = C$$

$$T_{front} = 0$$

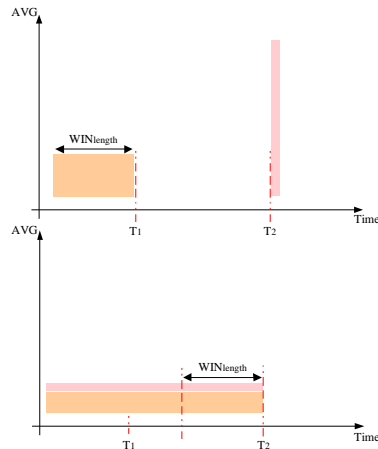
After every packet arrival:

$$Bytes_{TSW} = avg(n) \cdot Win_{length}$$

$$New_{bytes} = Bytes_{TSW} + PacketLength$$

$$avg(n+1) = \frac{New_{bytes}}{T_{now} - T_{front} + Win_{length}}$$

$$T_{front} = T_{now}$$



Conformance algorithms

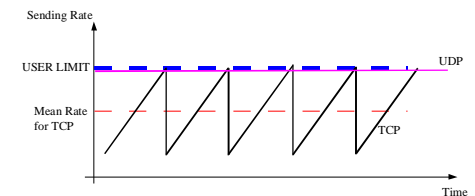
- Strict conformance
 - Packets exceeding contracted rate are marked immediately as non-conforming
- TSW conformance
 - Packets exceeding 1.33 times contracted rate are marked as non-conforming
- Probability conformance
 - Packets exceeding contracted rate are marked as non-conforming with increasing probability

Metering

- Based on the measured information a conformance statement is declared
- Conformance is the observation whether the measured variable is within predefined boundaries.
 - Customer has contracted rate of X bps with variation of x bps
 - Customer has contract of average rate X bps and peak of Y bps. He is allowed to send bursts of Z kB in peak rate.

Rate Control Problems

- Two parallel transport protocols with contradicting control:
 - UDP – with no control
 - TCP – with additive increase exponential decrease rate control
- Problem: Metering system cannot easily offer fair service to both TCP and UDP clients in the same system.



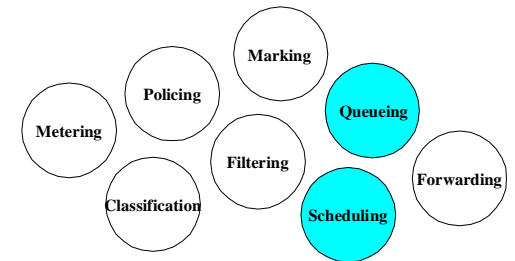
Marking

- Marker is used to attach conformance / class information to every packet.
- Marker uses IPv4 TOS/DSCP field to convey information for other processing elements in the network.
 - TOS
 - Prec: 3 bit priority
 - TOS: user preference for routing
 - DSCP
 - Class and precedence

Version	Header	TOS	Length
Ident		Flags	Offset
TTL	Protocol	Checksum	
SourceAddr			
DestinationAddr			
Options (variable)			PAD
Prec.		TOS	0

Output processor

- Output processor of a Internet router consists following elements
 - Queues and their management algorithms
 - Scheduling



Queues

- Queues are used to store **contending** packets
 - Contention is **temporary** event rising from statistical multiplexing
 - Packets from different input links of a router attempt to the same output link at certain time
 - Packets from a higher speed link arrive **temporarily** too fast for a slow speed link
- If contention is permanent queues overflow i.e. network is **congested**
- Difference:
 - Contention – packets are not lost only delayed
 - Congestion – packets are not only delayed but also lost

Queues

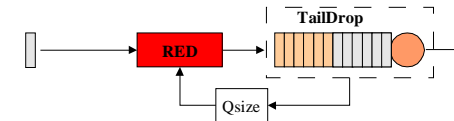
- Congestion situations demand **queue management** to decide
 - When packets should be discarded
 - Which are the packets that should be discarded
- Prevalent solutions
 - Tail Drop
 - Random Early Detection (RED)
 - Random Early Detection In/Out (RIO)

Tail Drop

- Simple algorithm:
 - If arriving packets sees a full queue it is discarded
 - Otherwise it is accepted to the queue
- Problem:
 - Poor fairness in distribution of buffer space
 - Unable to accommodate short transients when queue is almost full
 - Bursty discarding leading **global synchronisation**
- Global synchronisation is a process where large number of TCP connections synchronise their window control due to concurrent packet losses.
 - Packet losses are bursty, therefore window decreases to one and halts the communication

Random Early Detection

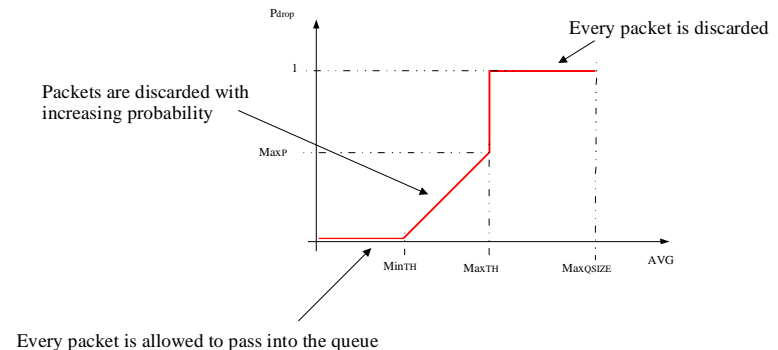
- RED is an active queue management algorithm (AQM), which aims to
 - Prevent global synchronisation
 - Offer better fairness among competing connections
 - Allow transient burst without packet loss
- Algorithm operates on the knowledge of current Qsize
 - Updated on every arrival and departure from the actual queue



RED

- Qsize is used to calculate average length of the queue:
 - Initial condition:**
 - $avg(0) = 0$
 - $Count = -1$
 - When Qsize=0:**
 - $T_{idle} = T_{now}$
 - After every packet arrival:**
 - if $Qsize(n) > 0$:
 - $avg(n+1) = (1-\epsilon) \cdot avg(n) + \epsilon \cdot Qsize(n)$
 - else:
 - $avg(n+1) = avg(n) \cdot (1-\epsilon)^{(T_{now}-T_{idle})}$
- If queue is empty, averaging is done based on the assumption that N packets have passed the algorithm before actual packet arrival. -> Decay of average during idle times**
- Packets are discarded based on the average queue length:
 - if $avg(n+1) < min_{th}$:
 - $Count = -1$
 - else if $min_{th} \leq avg(n+1) < max_{th}$:
 - $count = count + 1$
 - $P_b(n+1) = \max_p \frac{avg(n+1) - min_{th}}{max_{th} - min_{th}}$
 - $P_a(n+1) = \frac{P_b(n+1)}{1 - count \cdot P_b(n+1)}$
 - With probability $P_a(n+1)$:**
 - Discard packet
 - $Count = 0$
 - else if $max_{th} \leq avg(n+1)$:
 - Discard packet
 - $Count = 0$

RED



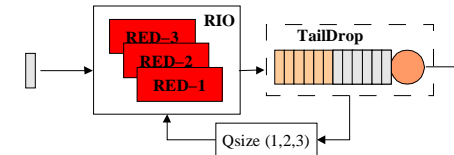
Every packet is allowed to pass into the queue

Achievements of RED

- Some packets are discarded even before overflow of the actual buffer
 - Is it good or bad ?
 - Bad: A part of buffer space is in some occasions wasted
 - Good: A signal is sent to co-operating sources that they should decrease their sending rate or congestion will occur
- On the average early packet discards will hit connections which use more than their fair share of capacity in contending link
 - Is it good or bad ?
 - Bad: Makes differentiation impossible
 - Good: Is consistent policy and withing the goal of conventional Best Effort model

RED In/Out – WRED

- When we aim for differentiation of resources we must also allow different shares of resources in contending link or buffer
- One way to do it is to use RED with several parallel algorithms and thresholds
 - RED In/Out → RIO or WRED
 - Popular implementations use two or three parallel algorithms
- This requires that packets are marked
 - One algorithm is responsible of one or several marks



RIO

- Operation is usually based on following idea:
 - Customer has contracted capacity of X bps
 - He sends packets with rate Y bps
 - If Y is greater than X, some packets are marked as out of profile.
 - Out of profile packets usually experience harsh treatment on contending situations
- Calculation of the average queue length is modified to take into account number of packets with different markings:
 - In (green): Only green packets
 - In/Out (yellow): Green and yellow packets
 - Out (red): All packets in the queue

Parameters in WRED

- All parameters are independent for different markings
 - More dimensions in creating differentiation
- Some parameters are common for different markings
 - Less dimensions but more understandable

