

Lecture 6.3

Transport Layer: Quality of Service

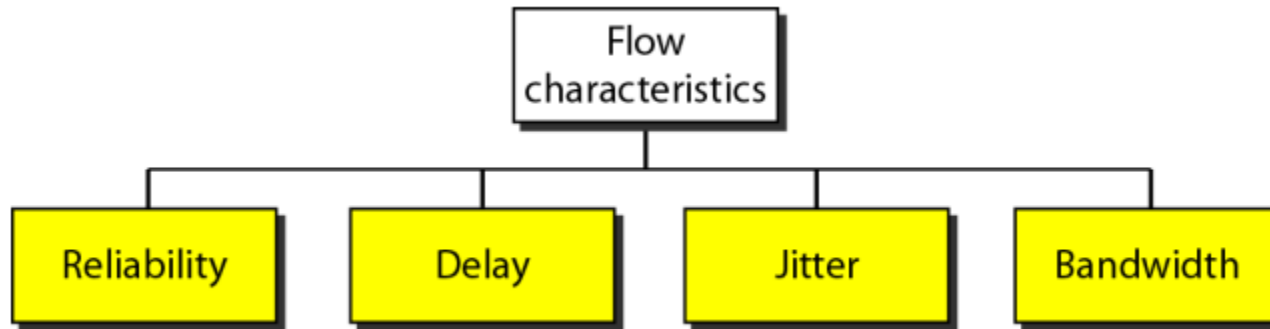
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Quality of Service

- **Quality of Service (QoS)** refers to the **capability of a network** to provide **better service** to **selected network traffic** over various technologies.
- The **primary goal** of **QoS** is to provide **priority** including dedicated **bandwidth**, **controlled jitter** and **latency** (required by some real-time and interactive traffic), and improved **loss characteristics**.
- One **important issue** is **making sure** that providing **priority** for one or more **flows** does not make other flows **fail**.
- A **flow** can be **defined** as a **combination** of **source** and **destination addresses**, **source** and **destination socket numbers** etc.

Flow Characteristics



1. Reliability

- **Reliability** is a **characteristic** that a **flow** needs.
- **Lack of reliability** means **losing a packet** or **acknowledgment**, which entails **retransmission**.
- However, the **sensitivity** of **application programs** to **reliability** is **not the same**.
- For **example**, it is **more important** that **electronic mail**, **file transfer** have **reliable transmissions** than **telephony** or **audio conferencing**.

Flow Characteristics

2. Delay

- Different applications can tolerate delay in different degrees.
- In this case, telephony, audio conferencing, video conferencing, and remote log-in need **minimum delay**, while delay in file transfer or e-mail is **less important**.

3. Jitter

- Jitter is the **variation in delay** for packets belonging to the same flow.
- For **example**, if four packets depart at times 0, 1, 2, 3 and arrive at 20, 21, 22, 23, all have the **same delay**, 20 units of time.
- On the other hand, if the above four packets arrive at 21, 23, 21, and 28, they will have **different delays**: 21, 22, 19, and 24.

Flow Characteristics

- For **applications** such as **audio** and **video**, the **first case** is **completely acceptable**; the **second case** is **not**.
- For **audio** and **video applications**, it does not matter if the packets arrive with a **short** or **long delay** as long as the **delay** is the **same** for **all packets**.
- **Jitter** is defined as the **variation in the packet delay**.
- **High jitter** means the **difference** between **delays** is **large**; **low jitter** means the **variation** is **small**.

4. Bandwidth

- Different **applications** need **different bandwidths**.
- In **video conferencing** we need to send **millions of bits per second** to refresh a **colour screen** while the total number of bits in an **e-mail** may not reach even a **million**.

TECHNIQUES TO IMPROVE QoS

- Some common methods to **improve QoS** are:

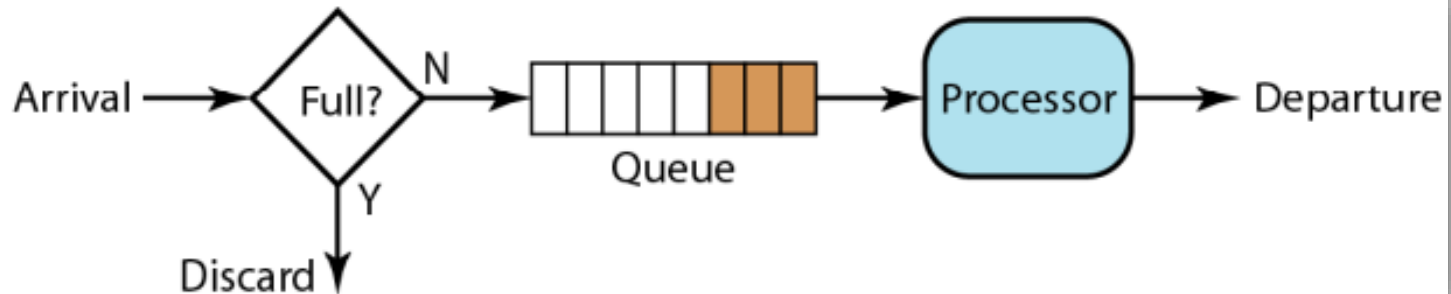
1. *Scheduling,*
2. *Traffic shaping,*
3. *Admission control,*
4. *Resource Reservation.*

1. Scheduling

- **Packets** from **different flows** arrive at a **switch** or **router** for **processing**.
- A **good scheduling technique** treats the **different flows** in a **fair** and **appropriate** manner.
- Several **scheduling techniques** are designed to **improve** the **quality of service**.
- The **three** of them here: **FIFO queuing**, **Priority queuing**, and **Weighted fair queuing(WFQ)**.

Scheduling: FIFO Queuing

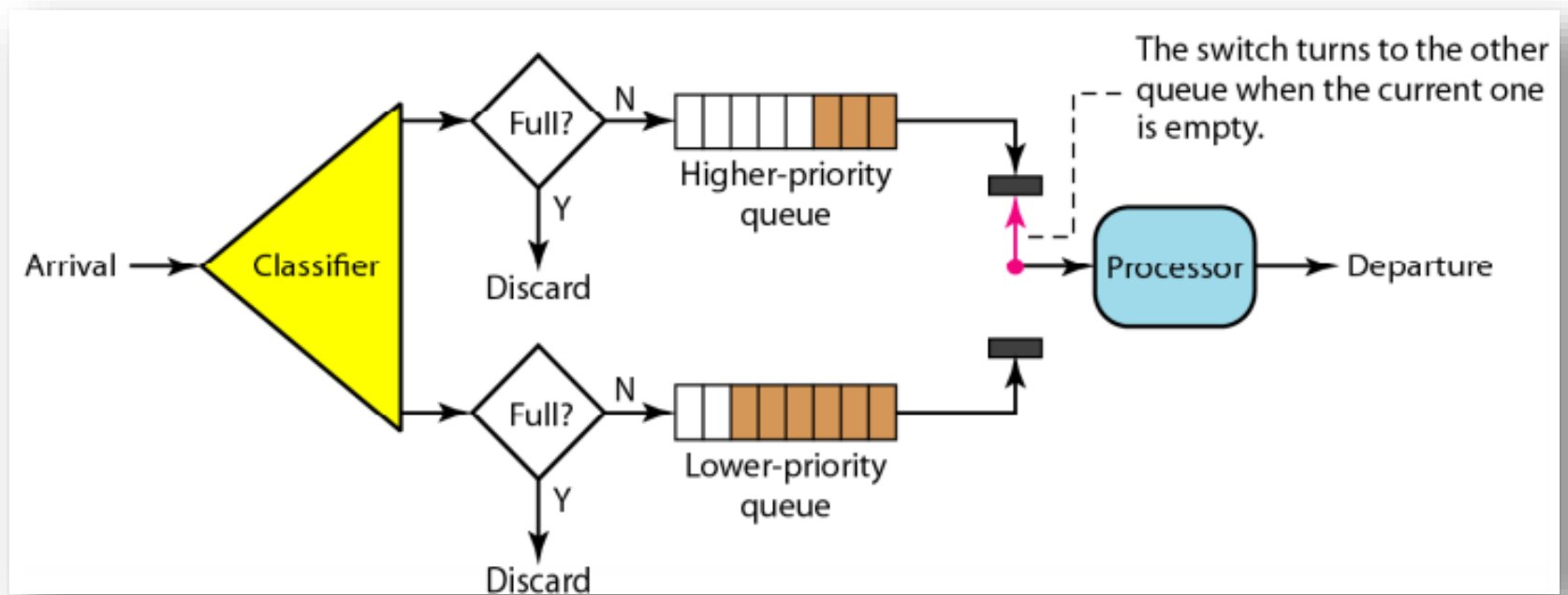
- In **first-in, first-out (FIFO) queuing**, packets wait in a **buffer (queue)** until the **node (router or switch)** is ready to **process** them.
- If the **average arrival rate** is **higher** than the **average processing rate**, the **queue** will **fill up** and **new packets** will be **discarded**.



Priority Queuing

- In **priority queuing**, packets are first assigned to a **priority class**.
- Each **priority class** has its **own queue**.
- The **packets** in the **highest-priority queue** are **processed first**.
- **Packets** in the **lowest-priority queue** are **processed last**.
- Note that the **system** does **not stop** serving a **queue** until it is **empty**.
- A **priority queue** can provide **better QoS** than the **FIFO queue** because **higher priority traffic**, such as **multimedia**, can reach the destination with **less delay**.
- However, there is a **potential drawback**. If there is a **continuous flow** in a **high-priority queue**, the **packets** in the **lower-priority queues** will **never have a chance to be processed**.
- This is a condition called **starvation**.

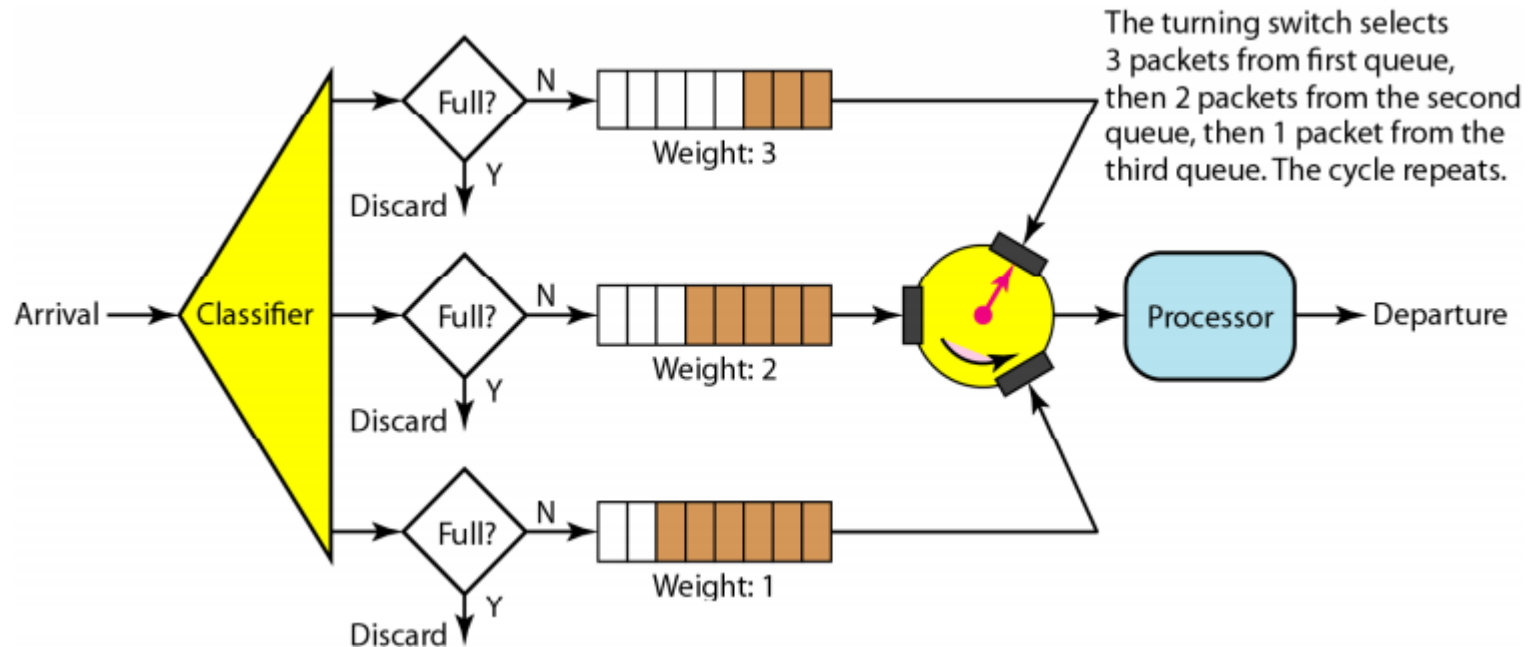
Priority Queuing



Weighted Fair Queuing

- In this **technique**, the **packets** are still assigned to **different classes** and **admitted** to **different queues**.
- The **queues**, however, are **weighted** based on the **priority** of the queues; **higher priority** means a **higher weight**.
- The **system** processes packets in each **queue** in a **round-robin fashion** with the **number of packets** selected from **each queue** based on the **corresponding weight**.
- For **example**, if the **weights** are **3, 2, and 1**, **three packets** are **processed** from the **first queue**, **two** from the **second queue**, and **one** from the **third queue**.
- If the **system** does not impose **priority** on the **classes**, all **weights** can be **equal**. In this way, we have **fair queuing** without priority.

Weighted Fair Queuing



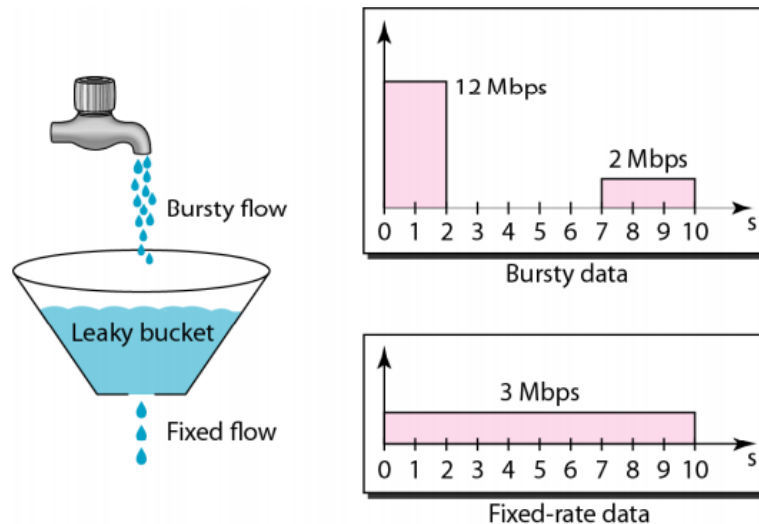
2.Traffic Shaping

- **Traffic shaping** is a mechanism to **control** the amount and the rate of the traffic sent to the network.
- Two techniques can shape traffic: **Leaky bucket** and **Token bucket**.

Leaky Bucket

- If a **bucket** has a **small hole** at the **bottom**, the **water leaks from the bucket** at a **constant rate** as long as there is **water** in the **bucket**.
- The **rate** at which the **water leaks** does not depend on the **rate** at which the **water is input** to the **bucket** unless the **bucket** is empty.
- The **input rate** can vary, but the **output rate** remains constant.
- Similarly, in **networking**, a technique called **leaky bucket** can **smooth out bursty traffic**.
- **Bursty chunks** are **stored** in the **bucket** and **sent out** at an **average rate**.

Leaky bucket



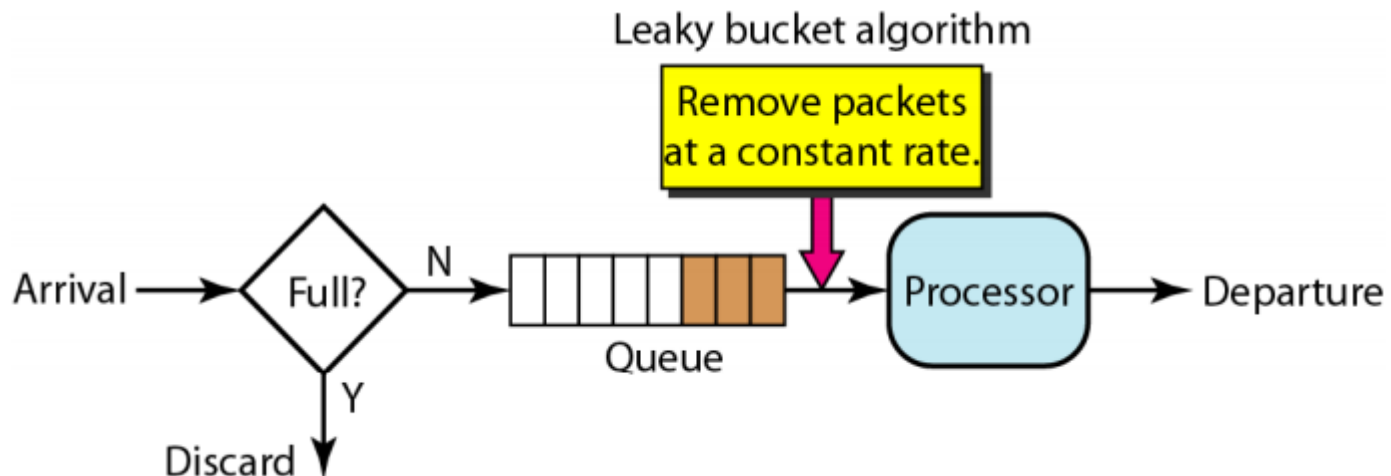
- Assume that the **network** has **committed** a **bandwidth of 3 Mbps** for a **host**.
- The use of the **leaky bucket** **shapes** the **input traffic** to make it conform to this **commitment**.
- The **host** sends a **burst of data** at a rate of **12 Mbps** for **2 s**, for a total of **24 Mbits** of **data**.
- The **host** is **silent** for **5 s** and then **sends data** at a rate of **2 Mbps** for **3 s**, for a total of **6 Mbits** of **data**.

Leaky bucket

- In all, the **host** has sent **30 Mbits** of **data** in all.
- The **leaky bucket smooths** the **traffic** by **sending out data** at a **rate** of **3 Mbps** during the same **10s**.
- **Without** the **leaky bucket**, the **beginning burst** may have **hurt the network** by **consuming more bandwidth** than is set aside for this host.
- We can also see that the **leaky bucket** may **prevent congestion**.

Leaky bucket implementation

- A simple **leaky bucket implementation** is shown in **Figure** below.
- A **FIFO queue** holds the **packets**.
- If the **traffic consists of fixed-size packets** (e.g., cells in ATM networks), the process **removes** a **fixed number** of **packets** from the **queue** at **each tick of the clock**.
- If the **traffic consists of variable-length packets**, the **fixed output rate** must be **based on the number of bytes or bits**.



Leaky bucket implementation

The following is an **Algorithm** for **variable-length packets**:

1. Initialize a *counter* to *n* at the tick of the clock as *counter=n*

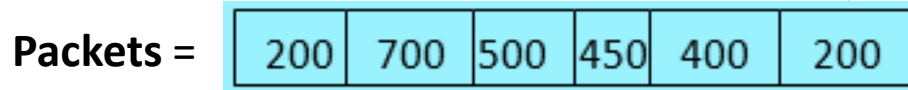
2. If *packet size < counter*, send the packet and decrement the *counter* by the *packet size* as *counter=n-packet size*

Repeat this step until *packet size > counter*.

3. Reset the *counter=n* and go to step 1.

Leaky bucket Example

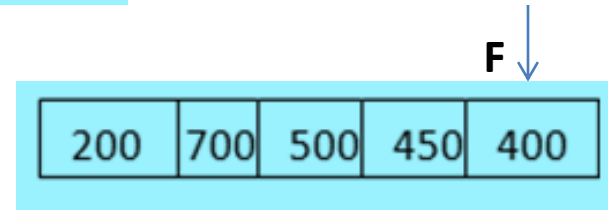
- Let $n=1000$



- Since $n > \text{front of Queue}$ i.e. $n > 200$

Therefore, $n = 1000 - 200 = 800$

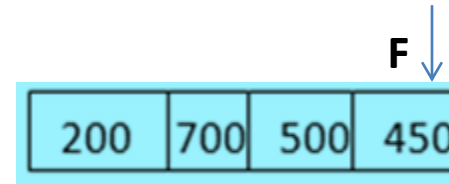
Packet of size **200** is sent to the network.



- Now Again $n > \text{front of the queue}$ i.e. $n > 400$

Therefore, $n = 800 - 400 = 400$

Packet of size **400** is sent to the network.



- Since $n < \text{front of queue}$

Therefore, the **procedure** is **stop**.

- Initialize $n=1000$ on another tick of clock.
- This **procedure** is repeated until all the packets are sent to the network.

Numerical: Leaky bucket

Question 1: Consider a host using **leaky bucket strategy** for traffic shaping. The host send a burst data at a rate of 15 Mbps for first 3 sec and remains silent for 2 sec. Then again a burst data at a rate of 6 Mbps is sent for next 2 sec and then the host remains silent for next 2 sec. Now again the host send data at a rate of 5 Mbps for next 3 sec. What will be the output data rate of the leaky bucket?

Solution:

Data sent is $15 \times 3 + 6 \times 2 + 5 \times 3 = 72$ Mb

During $3 + 2 + 2 + 2 + 3 = 12$ sec

So, the **output data rate** is $72/12 = 6$ Mbps

Numerical: Leaky bucket

Question 2: Consider a frame relay network having bucket capacity of **1Mb** and data is **input** at the **rate** of **25mbps**. Calculate

- i. What is the time needed to fill the bucket.
- ii. If the output rate is 2 mbps , the time needed to empty the bucket.

Solution:

Here , C is Capacity of bucket = 1Mb , Data input rate = 25 mbps , Output rate = 2mbps.

- i. $T = C / \text{input rate} = 1/25 = 40 \text{ msec}$
- ii. $T = C / \text{output rate} = 1/2 = 500 \text{ msec}$

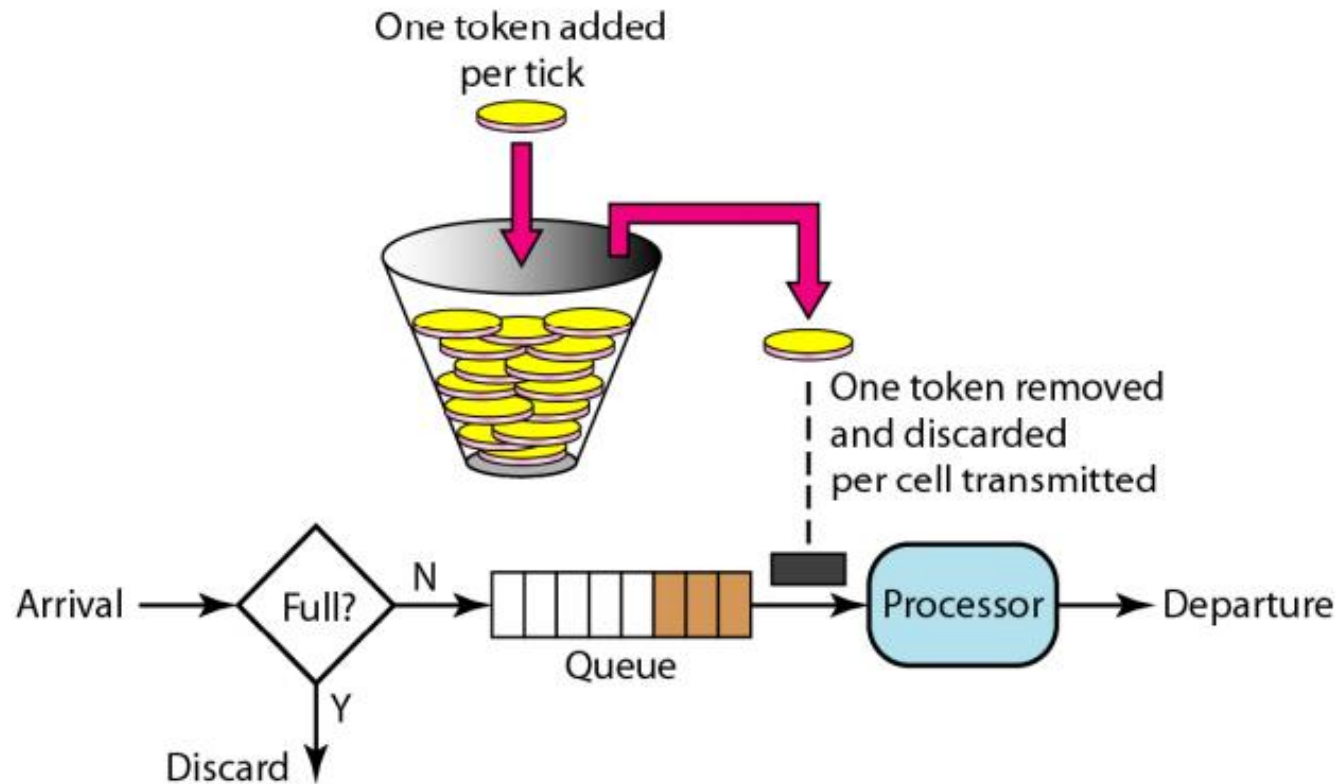
Token Bucket

- The **leaky bucket** is **very restrictive** as It does **not credit** an **idle host**.
- For **example**, if a **host is not sending for a while**, its bucket becomes **empty**.
- Now if the host has **bursty data**, the **leaky bucket** allows only an **average rate**.
- The **time when the host was idle** is not taken into account.
- On the other hand, the **token bucket algorithm** allows **idle hosts to accumulate credit** for the **future in the form of tokens**.
- For each tick of the **clock**, the system sends **n tokens to the bucket**.
- *The system removes **one token** for every byte of data sent.*
- For **example**, if **n is 100 and the host is idle for 100 ticks**, the bucket collects **10,000 tokens**.

Token Bucket

- Now the **host** can consume **all these tokens** in **one tick** with **10,000 bytes**, or the host takes **1000 ticks** with **10 bytes** per tick.
- In other words, the **host** can **send bursty data** as long as the **bucket** is **not empty**.
- The **token bucket** can easily be **implemented** with a **counter**.
- The **token** is **initialized** to **zero**.
- Each time a **token** is **added**, the **counter** is **incremented** by **1**.
- Each time a **unit of data** is **sent**, the **counter** is **decremented** by **1**.
- When the **counter** is **zero**, the **host** cannot send data.

Token Bucket



Token Bucket: Algorithm

1. A **token** is added to the bucket every **$1/r$ seconds**, i.e. token generation rate is **r tokens/second**.
2. The bucket can hold at the most **B tokens**. If a **token** arrives when the bucket is full, it is discarded.
3. When a **packet** of **n bytes** arrives,
 - if **at least n tokens** are in the bucket, **n tokens are removed** from the bucket, and the **packet is sent** to the network.
 - if **fewer** than **n tokens** are available, **no tokens are removed** from the bucket, and the **packet** is considered to be **non-conformant**.
 - The **non-conformant packets** may either be **dropped** or **queued** for subsequent transmission when **sufficient tokens** have **accumulated** in the bucket.

Token Bucket: Algorithm

- For these **non-conformant packets**, there are **two common options**:
- **Dropped (discarded)**: The packet is **removed immediately** to enforce the **traffic rate limit**.
- **Queued (delayed)**: The packet is **stored temporarily** until **enough tokens accumulate**, allowing it to be **sent later** without violating the rate limit.
- **Basic token update rule**
- Between events, **tokens increase linearly**:

$$T(t_{\text{new}}) = \min (B, T(t_{\text{old}}) + r \cdot \Delta t)$$

- When a **packet of size S** is transmitted, **tokens reduce**: **T = T-S**

Token Bucket: Example

***Question:** Consider the following parameters and packet arrival details. Find the packet processing sequence using Token Bucket.*

- **Parameters (units = bytes):**
 - Token generation rate $r=1000$ bytes/s
 - Bucket capacity $B=2000$ bytes
 - Initial tokens $T(0)=2000$ (bucket full)
- **Packet arrivals (time in seconds):**
 - **Packet 1** (P1): size 1500 bytes at $t=0.0$
 - **Packet 2** (P2): size 1200 bytes at $t=0.5$
 - **Packet 3** (P3): size 800 bytes at $t=1.0$

Token Bucket: Example

- **1) $t=0.0$: P1 arrives (1500 B)**
 - Initial tokens = 2000.
 - $2000 \geq 1500 \rightarrow$ **conformant** \rightarrow **send immediately.**
 - **Tokens after sending:** $2000 - 1500 = 500$
- **2) $t=0.5$: P2 arrives (1200 B)**
 - Time since last event = 0.5 s \rightarrow **tokens accumulate:** $r \cdot \Delta t = 1000 \times 0.5 = 500$
 - **Tokens before decision:** $\min(2000, 500 + 500) = 1000$
 - Need **1200**, but only **1000 available** \rightarrow **non-conformant** (short by 200 bytes).
- Now any one of **two policies** can be followed:

Token Bucket: Example

Policy A — Queue (delay until tokens available)


- Shortfall = $1200 - 1000 = 200$ bytes.
- With rate 1000 B/s, time to accumulate 200 bytes = $200/1000 = 0.2$ s.
- So P2 will be sent at $t = 0.5 + 0.2 = 0.7$ s.

At send moment $t = 0.7$:

- Tokens reached 1200, send P2 → tokens after sending = $1200 - 1200 = 0$.

Token Bucket: Example

Next, P3 at $t = 1.0$ (0.3 s after 0.7):

- Tokens accumulated in 0.3 s = $1000 \times 0.3 = 300$.
- Tokens available = $0 + 300 = 300$.
- P3 size 800 \rightarrow non-conformant (short by 500).
- Wait time to get 500 bytes = $500/1000 = 0.5$ s \rightarrow P3 sent at $t = 1.5$ s.
- After sending P3 at 1.5 s, tokens = 0. 

Token Bucket: Example

Policy B — Drop (discard non-conformant immediately)

- At $t = 0.5$, P2 is dropped. Tokens remain 1000.
- At $t = 1.0$ (0.5 s later), tokens accumulate: $1000 + 1000 \times 0.5 = 1500$ (below capacity).
- P3 arrives (800 B): $1500 \geq 800 \rightarrow$ **conformant**, send immediately.
- Tokens after sending = $1500 - 800 = 700$.

TECHNIQUES TO IMPROVE QoS

3. Resource Reservation

- A flow of data needs **resources** such as a **buffer, bandwidth, CPU time**, and so on.
- The **quality of service** is **improved** if these resources are **reserved beforehand**.
- There are some **QoS model** called **Integrated Services Model** and **Differentiated Services** , provide **resource reservation** to improve the quality of service.

4. Admission Control

- Admission control refers to the **mechanism** used by a **router**, or a switch, to **accept** or **reject a flow** based on **predefined parameters** called **flow specifications**.
- Before a **router accepts a flow** for **processing**, it checks the **flow specifications** to see if its **capacity** (in terms of bandwidth, buffer size, CPU speed, etc.) and its **previous commitments** to other flows can **handle the new flow**.