

$$T_f = \frac{L}{R}$$

$$T_{prop} = \frac{d}{v}$$

Stop and Wait ARQ

Efficiency

Efficiency (with Error)

$$P[\text{frame has no errors}] = (1-p)^n$$

$$P[\text{frame has errors}] = FER = P_e = 1 - (1-p)^n$$

$$C_{max} = R \times S$$

$$P[1 \text{ bit received in error}] = \binom{n}{1} p(1-p)^{n-1}$$

$$\mu = \frac{C}{L}$$

$$P[n \text{ bits received in error}] = \binom{n}{1} p(1-p)^{n-1}$$

λ - arrival cli
 μ - service ri
 ρ - traffic int

- $N = \lambda T$
 - » N - avera
 - » T - aver
 - » λ - arriv

- $T = T_w + T_s$
 - » T_w - tim
 - » T_s - serv

- $N_w = N_a + N_s$
 - » N_w - nu
 - » N_s - nur

- $N_w = \lambda T_w$

| | | | | | | | | | |
|-------------|-----|-----|-----|-----|-----|-----|-----|-----|-----|
| Subnet | 1 | 2 | 4 | 8 | 16 | 32 | 64 | 128 | 256 |
| Host | 256 | 128 | 64 | 32 | 16 | 8 | 4 | 2 | 1 |
| Subnet Mask | /24 | /25 | /26 | /27 | /28 | /29 | /30 | /31 | /32 |

Go-Back-N ARQ

$$W = M-1 = 2^k -1$$

Efficiency

Efficiency (with Error)

Selective Repeat ARQ

$$W = \frac{M}{2} = 2^{k-1}$$

Efficiency (with Error)

$$\text{If } W \geq 1+2a \rightarrow S=1$$

Poisson arrival, exponential service time

$$P(B) = \frac{(1-\rho)\rho^B}{1-\rho^{B+1}}$$

$$P(B) = \text{Probability W packet seen in B}$$

$$- B - \text{Number of buffers}$$

$$\text{If } \rho = 1, P(B) = \frac{1}{B+1}$$

$$\text{If } \rho \gg 1, P(B) = \frac{\rho-1}{\rho} \approx \frac{\mu-\lambda}{\lambda}$$

$M/D/1$
Poisson arrival, Deterministic (constant) service time $1/\mu$

$$S = \begin{cases} 1-p_e, & W \geq 1+2a \\ \frac{W(1-p_e)}{1+2a}, & W < 1+2a \end{cases}$$

Kendall Notation
A/S/s/K

A – arrival
S – service
s - number of servers
K – buffer capacity

$$S = \begin{cases} \frac{1-p_e}{1+2ap_e}, & W \geq 1+2a \\ \frac{W(1-p_e)}{(1+2a)(1-p_e+Wp_e)}, & W < 1+2a \end{cases}$$

$$T_f = \frac{L}{R}$$

$$T_{prop} = \frac{d}{v}$$

$$a = \frac{T_{prop}}{T_f}$$

$$C_{max} = R \times S$$

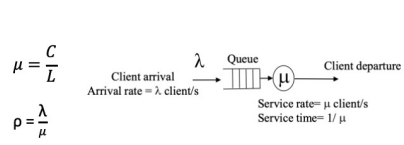
$$P[\text{frame has no errors}] = (1-p)^n$$

$$P[\text{frame has errors}] = FER = P_e = 1 - (1-p)^n$$

$$P[1 \text{ bit received in error}] = \binom{n}{1} p(1-p)^{n-1}$$

$$P[i \text{ bits received in error}] = \binom{n}{i} p^i (1-p)^{n-i}$$

$$\binom{n}{i} = \frac{n!}{(i!(n-i)!)}$$



λ - arrival client rate (client/s)
 μ - service rate (client/s)
 ρ - traffic intensity, how busy the server is (%)

- **N=λT**
 - » N - average number of clients in a system
 - » T – average amount of time a client spends in the system
 - » λ – arrival rate of clients to the system

- T=Tw+Ts
 - » Tw – time a client waits in the queue for being served
 - » Ts – service time

- N=Nw+N_s
 - » N_w – number of clients waiting in the queue for being served
 - » N_s – number of clients being served

- **N_w=λTw**

Stop and Wait ARQ

Efficiency

$$S = \frac{T_f}{T_{prop} + T_f + T_{prop}} = \frac{1}{1 + 2a}$$

Efficiency (with Error)

$$S = \frac{T_f}{E[A](T_f + 2T_{prop})} = \frac{1}{E[A](1 + 2a)} = \frac{1 - p_e}{1 + 2a}$$

M/M/1

Poisson arrival, exponential service time

$$T = N/\lambda = \frac{1}{\mu - \lambda}$$

$$T_w = T - T_s = \frac{1}{\mu - \lambda} - \frac{1}{\mu} = \frac{\rho}{\mu (1 - \rho)}$$

$$N_w = T_w \lambda = \frac{\lambda}{\mu - \lambda} - \frac{\lambda}{\mu} = N - \rho$$

$$N = \frac{\rho}{(1 - \rho)} = \frac{\lambda}{\mu - \lambda}$$

Probability of queue having n clients

$$P(n) = \rho^n (1 - \rho)$$

P(n) - probability of the Markov chain be in state n
 n - number of clients in the queue

Go-Back-N ARQ

$$W = M - 1 = 2^k - 1$$

Efficiency

$$\text{If } W \geq 1 + 2a \rightarrow S = 1$$

$$\text{If } W < 1 + 2a \rightarrow S = W / (1 + 2a)$$

Efficiency (with Error)

$$S = \begin{cases} \frac{1 - p_e}{1 + 2ap_e} & , W \geq 1 + 2a \\ \frac{W(1 - p_e)}{(1 + 2a)(1 - p_e + Wp_e)} & , W < 1 + 2a \end{cases}$$

M/M/1/B

Poisson arrival, exponential service time

$$P(B) = \frac{(1 - \rho) \rho^B}{1 - \rho^{B+1}}$$

- P(B) - Probability of packet being lost
- B - Number of buffers

$$\text{If } \rho = 1, P(B) = \frac{1}{B + 1}$$

$$\text{If } \rho \gg 1, P(B) = \frac{\rho - 1}{\rho} = \frac{\mu - \lambda}{\lambda}$$

Selective Repeat ARQ

$$W = \frac{M}{2} = 2^{k-1}$$

Efficiency (with Error)

$$S = \begin{cases} 1 - p_e & , W \geq 1 + 2a \\ \frac{W(1 - p_e)}{1 + 2a} & , W < 1 + 2a \end{cases}$$

M/D/1

Poisson arrival, deterministic (constant) service time 1/μ

$$T_w = \frac{\lambda}{2\mu^2 (1 - \rho)} = \frac{\rho}{2\mu (1 - \rho)}$$

Kendall Notation A/S/s/K

A – arrival
 S – service
 s - number of servers
 K – buffer capacity

| Subnet | 1 | 2 | 4 | 8 | 16 | 32 | 64 | 128 | 256 |
|-------------|-----|-----|-----|-----|-----|-----|-----|-----|-----|
| Host | 256 | 128 | 64 | 32 | 16 | 8 | 4 | 2 | 1 |
| Subnet Mask | /24 | /25 | /26 | /27 | /28 | /29 | /30 | /31 | /32 |

- Signals should be reconstructed on the receiver side

- **Nyquist** showed that a signal having a bandwidth of B Hz, can be fully reconstructed, with a sampling rate of 2B sample/s

$$C = 2B \log_2(M)$$

- 2B - channel capacity baudrate (symbol/s or baud)

- M - levels are encoded, M = 4 (-5V(00) , -2V(01) , 2V(10), 5V(11))

- Sampling at a higher rate does not bring additional information

Baseband transmission

- signal has frequencies from zero up to a maximum BH

Passband transmission

- signal uses band of frequencies around the frequency of the carrier f_c

- M-QAM(Quadrature Amplitude Modulation), M = 4

- In a stable M/D/1, μ = λ

Shannon's Theorem

- The maximum capacity of a channel is said to be:

$$C = B_c \log_2 \left(1 + \frac{P_r}{N_0 B_c} \right)$$

- B_c - bandwidth of the channel (Hz), sampling rate

- P_r - Signal power as seen by the receiver (W)

- N₀B_c - noise power within the bandwidth B_c, seen by the receiver (W)

- N₀ - White noise; noise power per unit bandwidth (W/Hz)

which can be also be expressed as:

$$C = B \log_2 \left(1 + \frac{S}{N} \right)$$

- where, S/N is the Signal to Noise Ratio

- SNR = Pr (dB) - N (dB)

If a bandpass channel has a bandwidth B_c= 100 kHz

and Signal to Noise ratio (SNR) at the receiver is

$$\gg P_r/(N_0 B_c) = 7 \rightarrow C = 100k \log_2 (1 + 7) = 300kbit/s$$

$$\gg P_r/(N_0 B_c) = 255 \rightarrow C = 100k \log_2 (1 + 255) = 800kbit/s$$

Power expressed in **W**, **dBW**, or **dBm**

$$\gg P_{dBW} = 10 \log_{10} P : P = 100mW \rightarrow P_{dBW} = 10 \log_{10}(100 \cdot 10^{-3}) = -10 \text{ dBW}$$

$$\gg P_{dBm} = 10 \log_{10}(P/1mW) : P = 100mW \rightarrow P_{dBm} = 10 \log_{10}(100) = 20 \text{ dBm}$$

Internet transport-layer protocols

TCP service:

- **connection-oriented**: setup required between client and server processes

- **reliable transport** between sending and receiving process

- **flow control**: sender won't overwhelm receiver

- **congestion control**: throttle sender when network overloaded

- does not provide: timing, minimum throughput guarantees, security

UDP - User Datagram Protocol (UDP)

- Allows applications

- to interface directly to IP
- with minimal additional protocol overhead

- often used for streaming multimedia apps

- loss tolerant
- rate sensitive

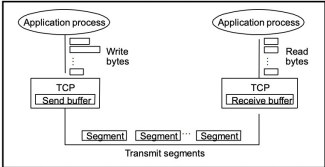
- other UDP uses
- DNS
- SNMP

- reliable transfer over UDP: add reliability at application layer
- application-specific error recovery!

TCP – Transmission Control Protocol

- **full duplex data**:
 - bi-directional data flow in same connection
- **connection-oriented**:
 - handshaking (exchange of control msgs) inits sender, receiver state before data exchange
- **flow controlled**:
 - sender will not overwhelm receiver
 - ARQ mechanism
- **Congestion control**
 - Avoids network's congestion
- **point-to-point**:
 - one sender, one receiver

- **reliable, in-order byte stream**:
 - no "message boundaries"
- **pipelined**:
 - TCP congestion and flow control set window size
- **send & receive buffers**



Packets may be lost

- Receiver confirms (ACK) reception of each segment

- If sender does not receive confirmation within a time limit → re-sends segment

Packets may contain errors

- Detected using checksum

- Packets with errors are ignored → handled as lost packets

- RTT → Round Trip Time

- sampleRTT measured for each segment/ACK pair

- Moving average (smooth RTT) of SRTT

- $SRTT_{new} = (1 - \alpha) \times SRTT_{old} + \alpha \times sampleRTT$
 - $\alpha = 1/8$

- Variation of RTT

- $RTTVAR_{new} = (1 - \beta) \times RTTVAR_{old} + \beta \times |SRTT_{old} - sampleRTT|$
 - $\beta = \lambda$

- RTO = SRTT + max (G, K*RTTVar)

- clock granularity of G seconds

- K = 4

- RTO - retransmission timeout

MAC

Datalink has 2 sub-layers:

- LLC (Logical Link Control)
 - interface for network layer, error and flow control
 - Error and flow control
- MAC (Medium Access Control)
 - Access control to the shared medium
 - Frame transmission/reception
 - Addressing and Error control

ideal Multiple Access Protocol

- one station wants to transmit -> it uses the R bit/s
- m stations want to transmit -> each station uses an average rate R/m bit/s

MAC Protocols:

- Channel Partitioning
- Time Division Multiplexing | Frequency Division Multiplexing
- Random Access
 - channel not divided, collisions allowed
- Taking turns
 - stations take turns
 - stations with more data to send can take longer turns

Channel partitioning

MAC protocols (TDM, FDM)

- » share channel efficiently and fairly at high loads
- » are inefficient at low loads
 - delay in channel access; 1/N bandwidth allocated even if only 1 active node!

Random access MAC protocols (Aloha, CSMA, CSMA/CD, CSMA/CA)

- » efficient at low load -> single node can fully utilize the channel
- » high load -> collisions -> inefficiency
- Taking turns protocols
- » look for best of both worlds!

What Are Random Access Protocols?

- When station has packet to send
 - transmits at channel data rate R bit/s
 - no a priori coordination among stations
- If two or more stations transmit simultaneously → collision
- It defines:
 - when to send data
 - how to detect collision
 - how to recover from collisions
- Examples:
 - ALOHA, CSMA, CSMA/CD, CSMA/CA

1. ALOHA

- When packet transmission fails, it will delay a random period of time before next transmission
- Pure Aloha (unslotted)
 - Station transmits when it has a frame to transmit
 - $S_{max} = 18.4\%$
- Slotted Aloha
 - Time divided into time slots
 - $T_{slot} = T_{frame}$
 - (Re)transmissions only the beginning of a slot
 - $S_{max} = 36.8\%$

2. CSMA (Carrier Sense Multiple Access)

- Listens before transmit
 - If channel sensed free → transmit frame
 - If channel sensed busy → defer transmission
- Collisions can still occur (entire packet is lost)
- vulnerability time = $2 * T_{prop} = rtt$ (round trip time)
- Determine collision probability
- $a = T_{prop} / T_{frame} << 1$

3. CSMA/CD – Carrier Sense Multiple Access / Collision Detection

- Carrier Sense
 - station senses medium before transmitting
 - If free → station starts transmission
 - If busy → waits until free and then transmits
- Collision Detection
 - station listens medium while transmitting
 - if collision is detected
 - transmission is aborted
 - retransmission delayed using
 - a Binary Exponential Back-off algorithm
 - time modeled in time slots; $T_{slot} = 2 * T_{prop}$
 - after the ith consecutive collision → the station attempts to transmit, after waiting, a random number of slots uniformly distributed in $[0, 2^i - 1]$
- No ACK is needed
- It can detect itself for collisions
- however, a minimum frame size is required for collision detection

4. CSMA/CA - CSMA with Collision Avoidance

- CSMA/CD cannot be used for wireless transmission
- Therefore it is more commonly used in wireless networking
- CSMA / CA is effective before a collision.
- whereas, CSMA/CD is effective after a collision.
- Station with a frame to transmit
 - monitors the channel activity → until an idle period equal to a Distributed Inter-Frame Space (DIFS) has been observed
 - then if medium remains free → transmits frame
- If the medium is sensed busy
 - random backoff interval is selected
 - backoff time counter is decremented as long as the channel is sensed idle
 - stopped when a transmission is detected on the channel
 - reactivated when the channel is sensed idle again for more than a DIFS
 - the station transmits when the backoff time reaches 0
- to avoid channel capture
 - station waits random backoff time between two consecutive frame transmissions
 - even if the medium is sensed idle in the DIFS time
- ACK will be required for CSMA/CA
 - Since it cannot locally detect if there is a collision
- $S_{max} = 60\%$ commonly found

Taking turns Mac Protocols

1. Polling

- master station invites slave stations to transmit in turn
- concerns
 - polling overhead
 - latency
 - single point of failure (master)

2. Token Passing

- » control token
- passed from one station to next sequentially
- » token message
- » concerns
 - token overhead
 - latency
 - single point of failure (token)

Hub

- Has repeaters, repeats bits received on one port to all other ports
- Layer 1 device (does not know addresses)
- if collision detected on one port -> repeats random bits on other port
- 1 Collision Domain and half duplex

Bridge

- » Layer 2 device
- » forwards MAC frames to destinations based on MAC addresses
- » Packet received on one port -> analyzed by bridge -> re-sent on some other port
- » Bridge separates collision domains
 - a bridged LAN maybe larger than a repeated LAN
 - several frames may be transmitted simultaneously

Note: Switches not need a MAC Address (transparent bridging), but routers do have.

Services provided by network layer

Virtual Circuit network

- connection-oriented service

Datagram network

- connectionless service

Virtual Circuit

1. circuit establishment
2. data transference
3. circuit termination

- Packet carries identifier of Virtual Circuit

- Path defined from source to destination

- sequence of VC identifiers, one for each link along path

- Router

- maintains “state” for every supported circuit

- may allocate resources (bandwidth, buffers) per Virtual Circuit

Datagram

- No circuit establishment; no circuit concept

- Packets

- forwarded using destination host address

- packets between same source-destination pair may follow different paths

| Issue | Datagram subnet | Virtual-circuit subnet |
|---------------------------|--|--|
| Circuit setup | Not needed | Required |
| Addressing | Each packet contains the full source and destination address | Each packet contains a short VC number |
| State information | Routers do not hold state information about connections | Each VC requires router table space per connection |
| Routing | Each packet is routed independently | Route chosen when VC is set up; all packets follow it |
| Effect of router failures | None, except for packets lost during the crash | All VCs that passed through the failed router are terminated |
| Quality of service | Difficult | Easy if enough resources can be allocated in advance for each VC |
| Congestion control | Difficult | Easy if enough resources can be allocated in advance for each VC |

What is ARP needed for?

- On the same LAN you need to connect on the link layer
 - Known: IP address of destination (packet from Network Layer)
 - Unknown: MAC address of destination

- Protocol used to obtain the MAC address associated to a given IP address

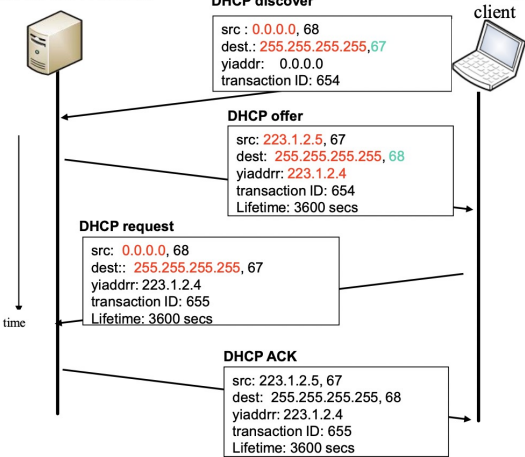
- RARP – Reverse ARP - Protocol used to obtain the IP address associated to a MAC address

DHCP allows

- host to dynamically obtain its IP address (assigns a dynamic IP address) from network server when it joins network

- It supports address reuse

DHCP server: 223.1.2.5



NAT

- To overcome IP address shortage
- Uses private IP address
- Converts an internal private IP address and port to a public one, to be used in the public network

ICMP - Internet Control Message Protocol

- Ping request
- Used by router or host
 - to send layer 3 error or control messages
 - to other hosts or routers
- Carried in IP datagrams

• Link-Local

- Used for communication between hosts in the same LAN/link
- Address built from MAC address
- Routers do not forward packets having Link-Local destination addresses

• Global Unicast

- Global addresses
- Address: network prefix + computer identifier
- Structured prefixes
- Network aggregation; less entries in the router forwarding tables

• Anycast

- Group address; packet is received by any (only one) member of the group

• Multicast

- Group address; packet received by all the members of the group