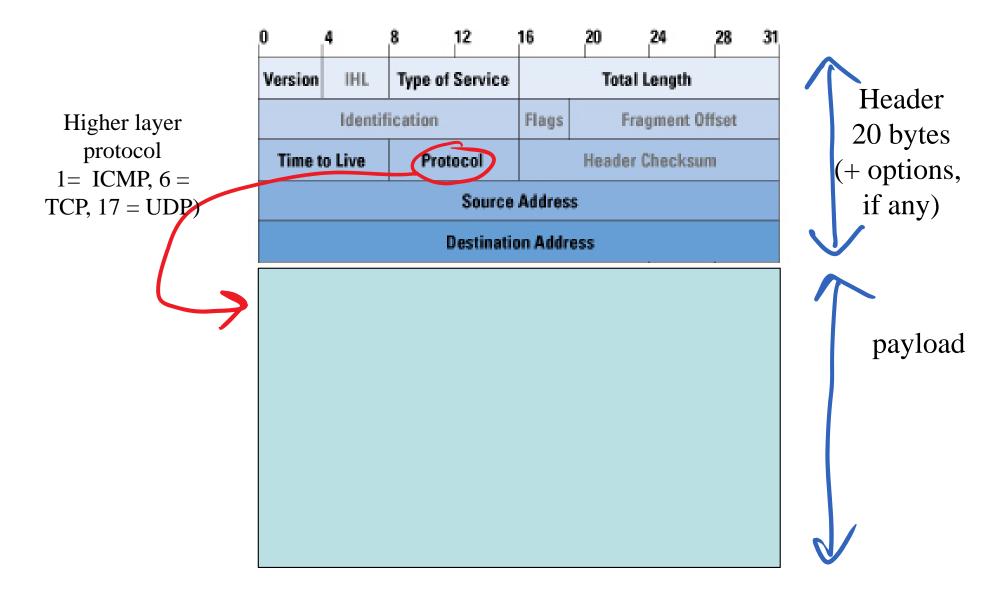
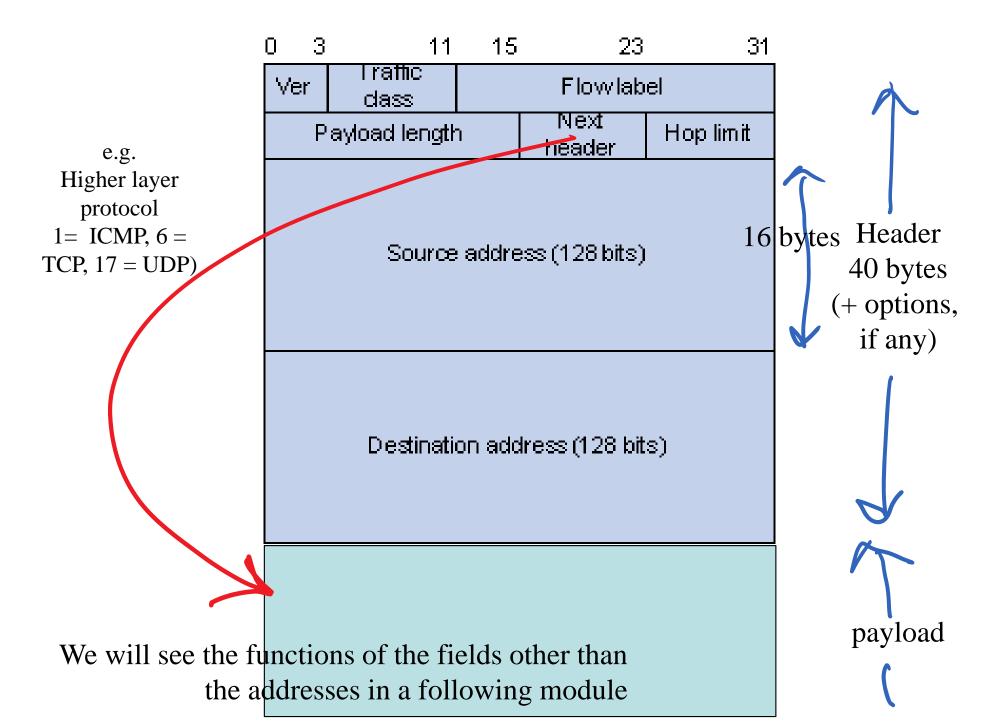


IPv4 Packet Format



We will see the functions of the fields other than the addresses in a following module

IPv6 Packet Format



Ethernet Frame format

Ethernet frame = Ethernet PDU

An Ethernet frame typically transports an IP packet, sometimes also other

Type of protocol contained in the Ethernet packet (hexa):

0800: IPv4

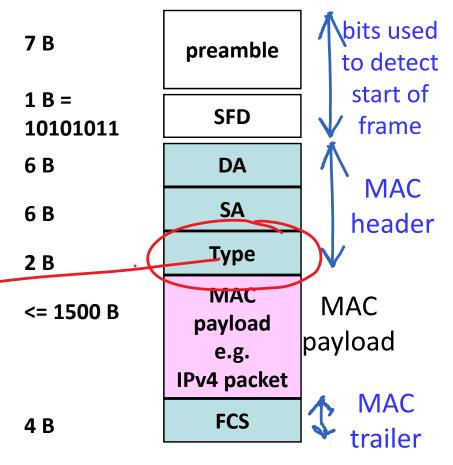
0806: ARP (used by IPv4)

86DD: IPv6

8847: MPLS unicast

88F7: Precision Time Protocol





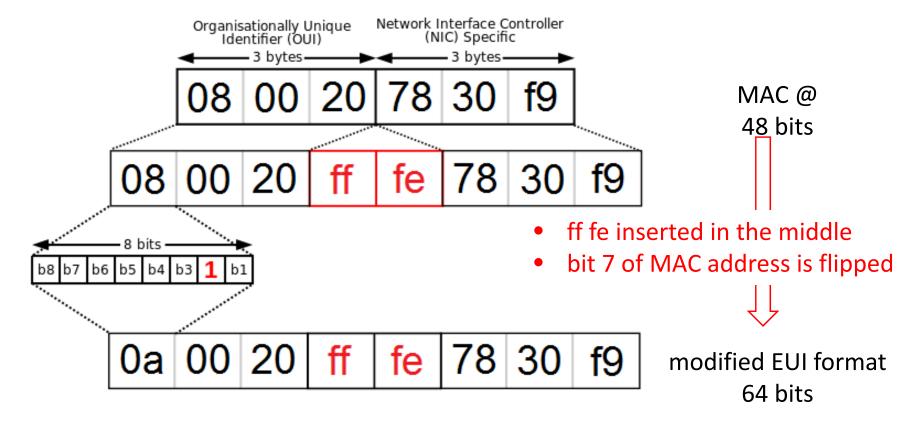
DA = destination address
SA = source address

Multicast MAC Addresses

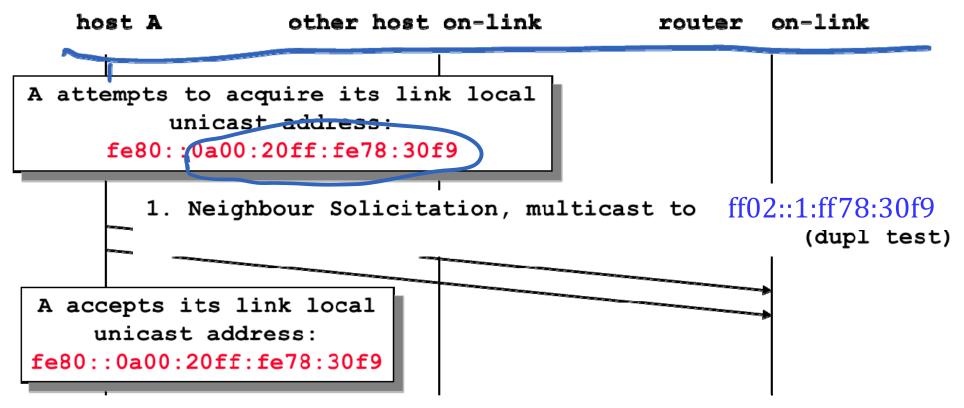
MAC multicast addr.	Used for
01-00-5e-XX-XX	IPv4 multicast
33-33-XX-XX-XX	IPv6 multicast

IP dest address	229.130.54.207
IP dest address (hexa)	e5-82-36-cf
IP dest address (bin)	10000010
Keep last 23 bits (bin)	00000010
Keep last 23 bits (hexa)	02-36-cf
MAC address	01-00-5e-03-36-cf

Host Part derived from MAC address: MAC@ → EUI (Extended Unique Identifier)



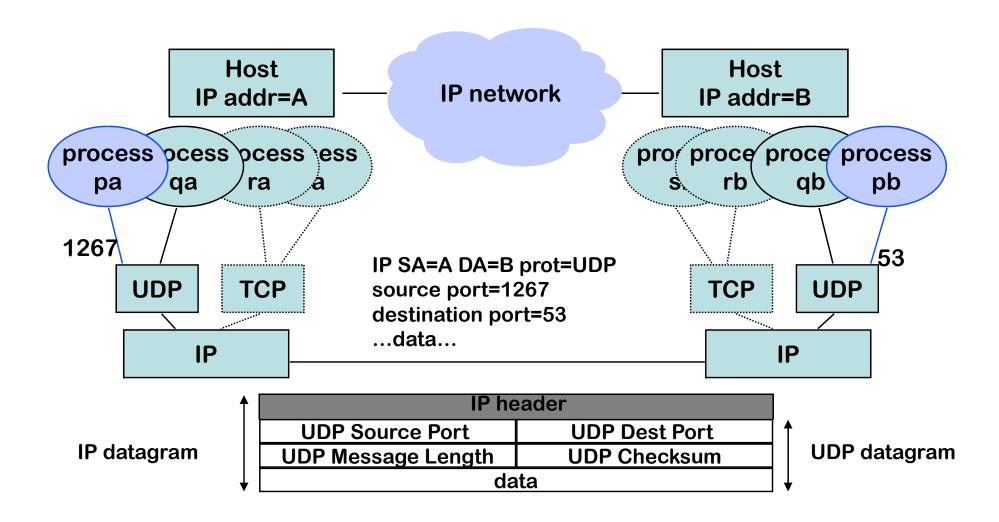
SLAAC Step 2: Duplicate Test



A sends a Neighbour Solication (NS) message to check for address duplication, sent to the Solicited Node Multicast Address.

Any host that would have to same link local address listens to this multicast address

UDP Uses Port Numbers



```
IP header (20 or 40 B + options)
                         dest port
    srce port
            sequence number
              ack number
                                           TCP
hlen
             flags
                          window
      rsvd
                                           header
                                           (20 Bytes +
     checksum
                      urgent pointer
                                           options)
      options (SACK, ...)
                               padding
                                           <= MSS bytes
         segment data (if any)
```

flags	meaning
NS	used for explicit congestion notification
CWR	used for explicit congestion notification
ECN	used for explicit congestion notification
urg	urgent ptr is valid
ack	ack field is valid
psh	this seg requests a push
rst	reset the connection
syn	connection setup
fin	sender has reached end of byte stream

Dijkstra's Shortest Path Algorithm

The nodes are 0...N; the algorithm computes shortest paths from node 0.

c(i,j): cost of link (i,j).

```
m(0) = 0; m(i) = \infty \ \forall \ i \neq 0; V = \emptyset; pred(i) = \emptyset \ \forall i;
for k = 0: N do
         find i \notin V that minimizes m(i)
         if m(i) is finite
                   add i to V
                  for all neighbours j \notin V of i
                            if m(i) + c(i, j) < m(j)
                                     m(j) = m(i) + c(i, j)
                                     pred(j) = \{i\}
                            else if m(i) + c(i, j) = m(j)
                                     m(j) = m(i) + c(i, j)
                                     pred(j) = pred(j) \cup \{i\}
```

V: set of nodes visited so far.

pred(i): estimated set of predecessors of node i along a shortest path (multiple shortest paths are possible).

m(j): estimated distance from node 0 to node j.

At completion, m(i) is the true distance from 0 to i.

Practical Aspects

OSPF packet are sent directly over IP (OSPF=protocol 89 (0x59)). Reliable transmission is managed by OSPF with OSPF Acks and timers.

OSPFv2 supports IPv4 only OSPFv3 supports IPv6 and dual-stack networks

OSPF routers are identified by a 32 bit number OSPF areas are identified by a 32 bit number

The Centralized Bellman-Ford Algorithm

Algorithm BF-C

input: a directed graph with links costs A(i,j); assume A(i,j)>0 and $A(i,j)=\infty$ when nodes i and j are not connected.

output: vector p s.t. p(i)= cost of best path from node i to node 1

$$p^{0}(1) = 0, \quad p^{0}(i) = \infty \text{ for } i \neq 1$$

for $k = 1, 2, ...$ do
$$p^{k}(i) = \min_{j \neq i} [A(i, j) + p^{k-1}(j)] \text{ for } i \neq 1$$

$$p^{k}(1) = 0$$
until $p^{k} = p^{k-1}$
return (p^{k})

Distributed Bellman-Ford

Requires only to remember distance from self to destination + the best neighbor (nextHop(i))

and works for all initial conditions

Distributed Bellman-Ford Algorithm, BF-D

```
node i maintains an estimate q(i) of the distance p(i) to node 1; node i remembers the best neighbor nextHop(i)
```

initial conditions are arbitrary but q(1) = 0 at all steps;

from time to time, i sends its value q(i) to all neighbors

when *i* receives an updated value q(j) from *j*, node *i* recomputes q(i): eq(2) if j == nextHop(i) $then q(i) \leftarrow A(i,j) + q(j)$ $else q(i) \leftarrow min(A(i,j) + q(j), q(i))$

if eq(2) causes q(i) to be modified, nextHop $(i) \leftarrow j$

The BGP Decision Process

The decision process decides which route is selected;

At most one best route to exactly the same prefix is chosen

Only one route to 2.2/16 can be chosen

But there can be different routes to 2.2.2/24 and 2.2/16

A route can be selected only if its next-hop is reachable

Routes are compared against each other using a sequence of criteria, until only one route remains. The default sequence is

- O. Highest weight (Cisco proprietary)
- 1. Highest LOCAL-PREF
- 2. Shortest AS-PATH
- 3. Lowest MED, if taken seriously by this network
- 4. E-BGP > I-BGP
- 5. Shortest path to NEXT-HOP, according to IGP
- 6. Lowest BGP identifier

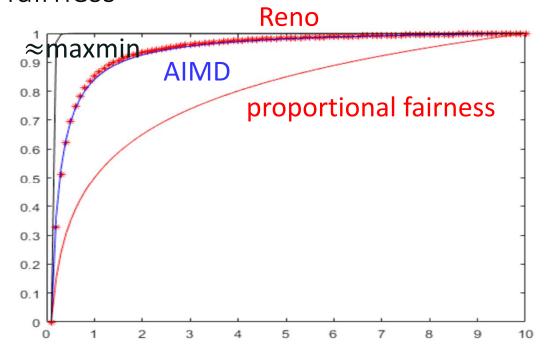
Fairness of TCP Reno

For long lived flows, the rates obtained with TCP are as if they were distributed according to utility fairness, with utility of flow i given by

$$U(x_i) = \frac{\sqrt{2}}{\tau_i} \arctan \frac{x_i \tau_i}{\sqrt{2}}$$

with $x_i = \text{rate} = W/\tau_i$, $\tau_i = \text{RTT}$

For sources that have same RTT, the fairness of TCP is between maxmin fairness and proportional fairness, closer to proportional fairness



rescaled utility functions; RTT = 100 ms maxmin approx. is $U(x)=1-x^{-5}$

TCP Reno

Loss - Throughput Formula

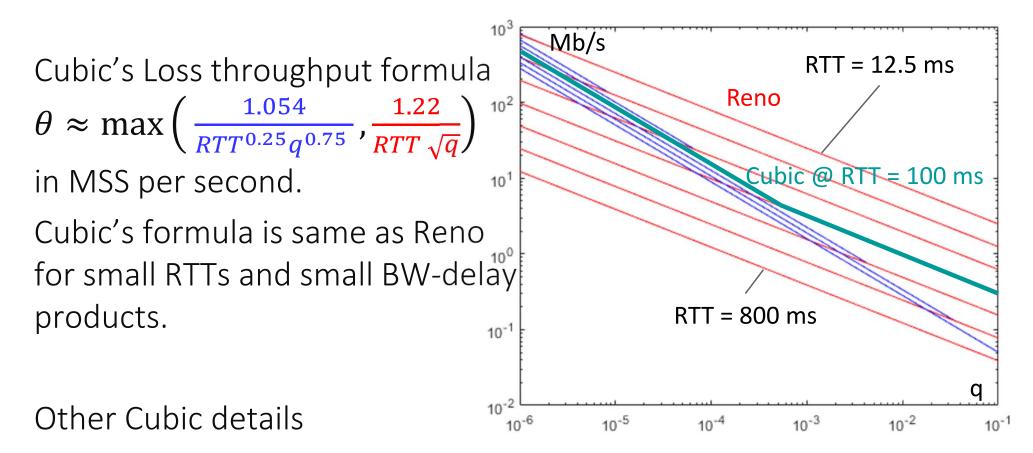
Consider a *large* TCP connection (many bytes to transmit)

Assume we observe that, in average, a fraction q of packets is lost (or marked with ECN)

The throughput should be close to
$$\theta = \frac{MSS\ 1.22}{RTT\ \sqrt{q}}$$

Formula assumes: transmission time negligible compared to RTT, losses are rare, time spent in Slow Start and Fast Recovery negligible, losses occur periodically

Cubic's Other Bells and Whistles



 W_{max} computation uses a more complex mechanism called "fast convergence"

see Latest IETF Cubic RFC / Internet Draft

or http://elixir.free-electrons.com/linux/latest/source/net/ipv4/tcp cubic.c