



Exam Booklet

TCP/IP Networking 2020-2021

Special Addresses

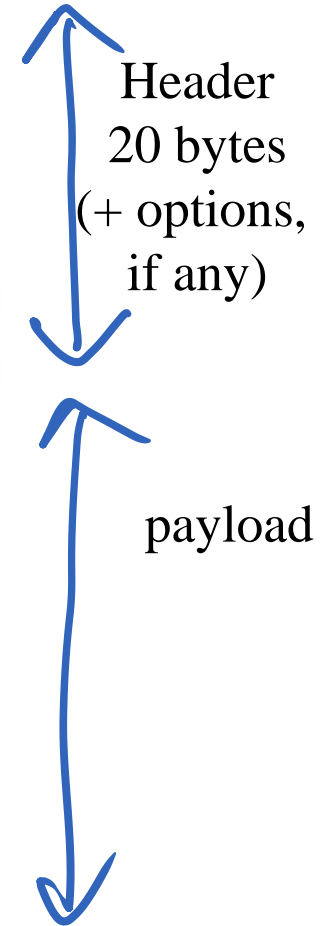
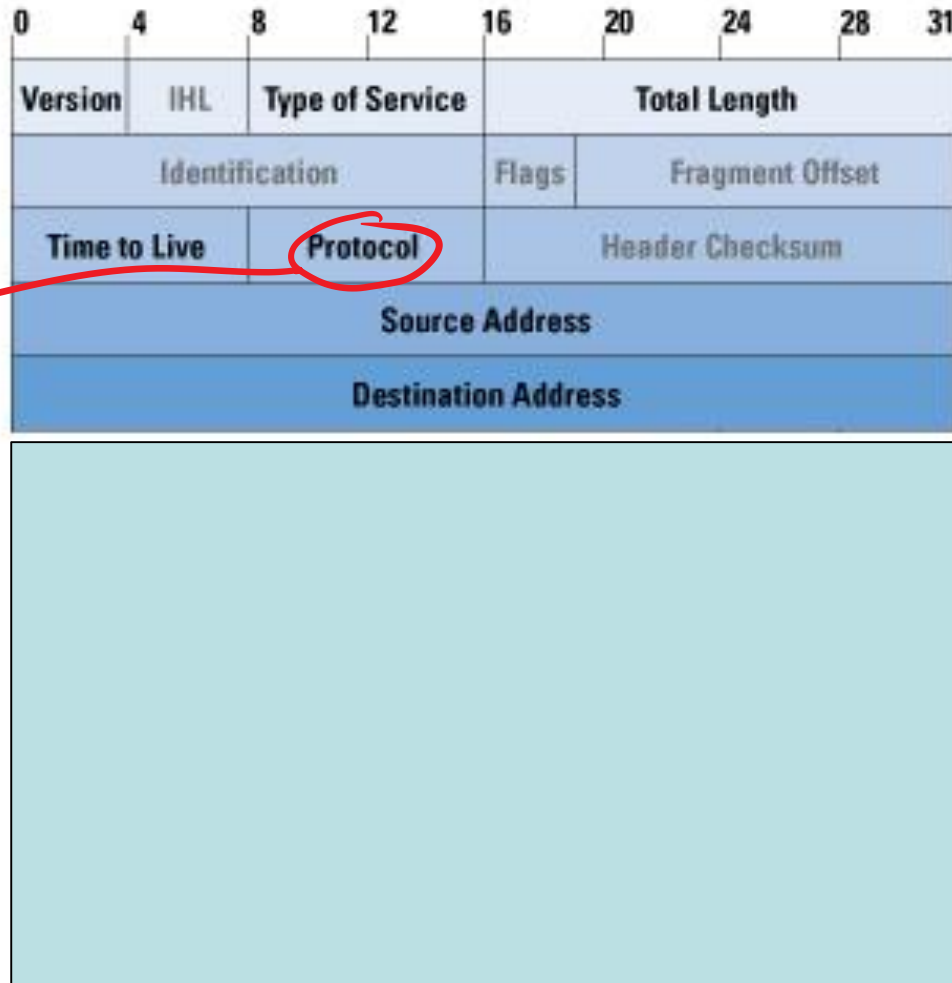
0.0.0.0	absence of address
127.0.0/24 for example 127.0.0.1	this host (loopback address)
10.0.0.0/8, 172.16.0.0/12, 192.168.0.0/16	private networks (e.g in IEW) cannot be used on the public Internet
169.254.0.0/16	link local address (can be used only between systems on same LAN)
224/4	multicast
240/5	reserved
255.255.255.255/32	link local broadcast

Examples of Special Addresses

	::/128	absence of address
	::1/128	this host (loopback address)
EPFL Private	fc00::/7 (i.e. fcxx: and fdxx:) For example fd24:ec43:12ca:1a6:a00: 20ff:fe78:30f9	Unique local addresses = private networks (e.g in IEW) cannot be used on the public Internet
	fe80::/10	link local address (can be used only between systems on same LAN)
	ff00::/8	multicast
	ff02::1:ff00:0/104	Solicited node multicast
	ff02::1/128	link local broadcast
	ff02::2/128	all link local routers

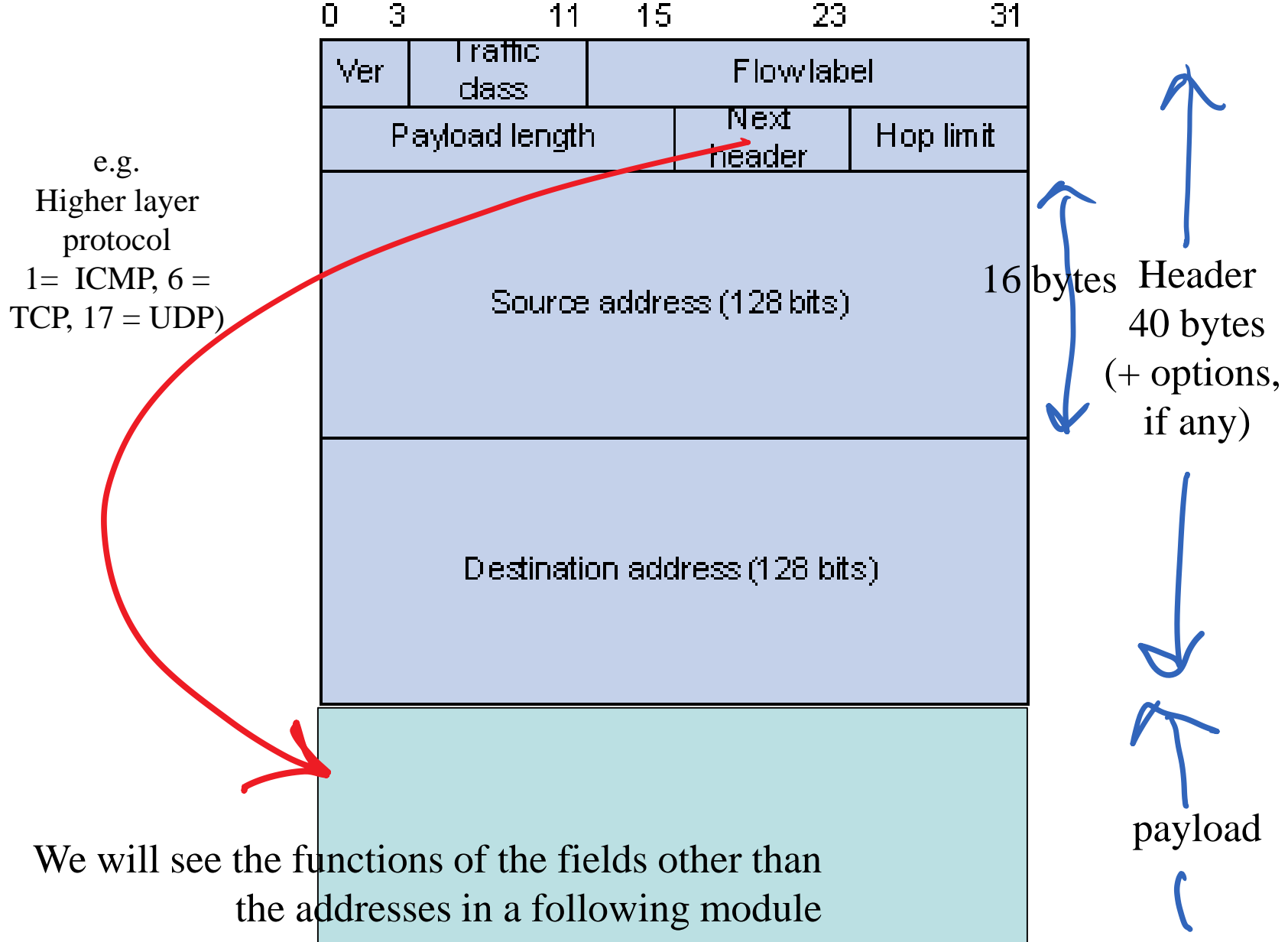
IPv4 Packet Format

Higher layer
protocol
1= ICMP, 6 =
TCP, 17 = UDP)



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

IPv6 Packet Format



Ethernet Frame format

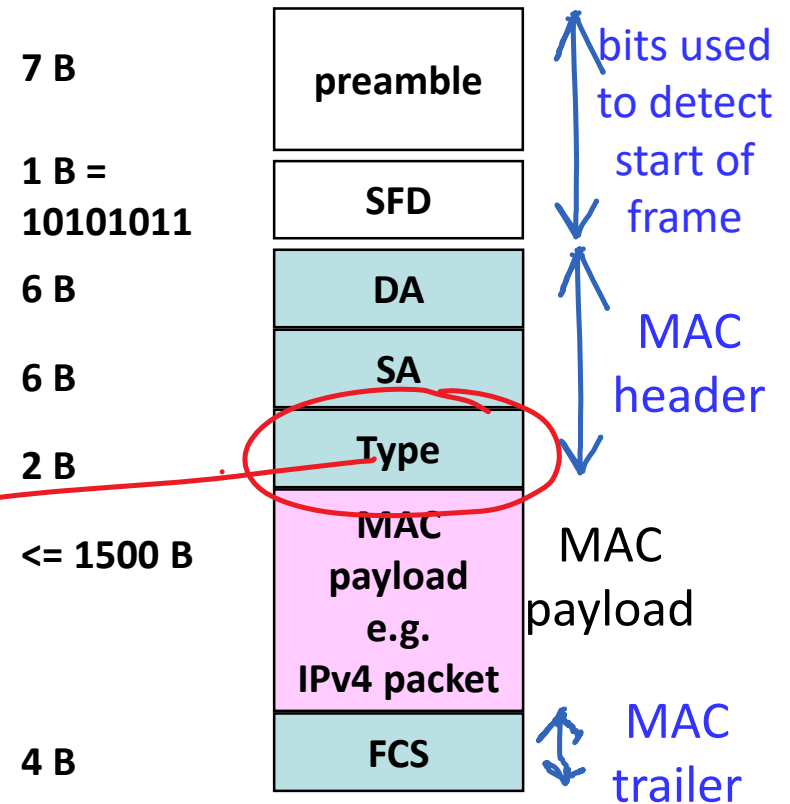
Ethernet frame = Ethernet PDU

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

Ethernet V.2 frame



DA = destination address
SA = source address

Multicast MAC Addresses

IP multicast address is **algorithmically** mapped to a multicast MAC address.

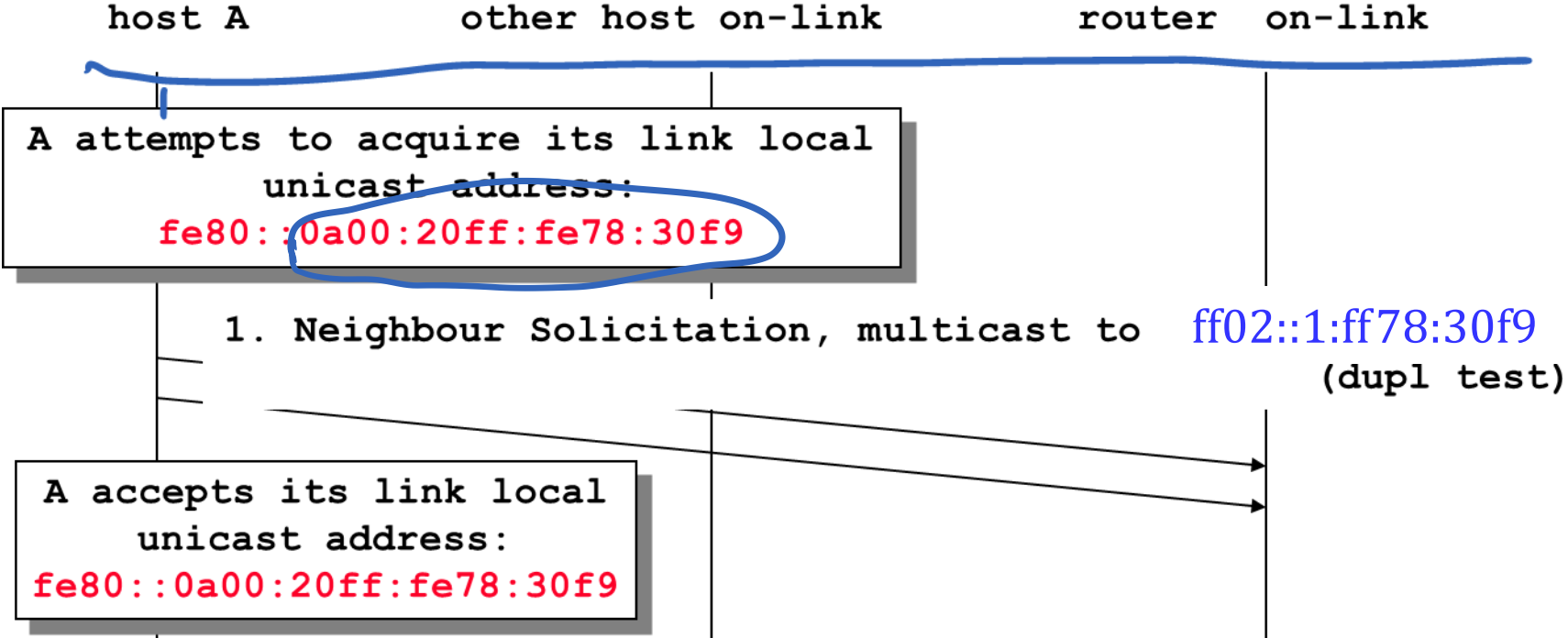
Last 23 bits of IPv4 multicast address are used in MAC address

Last 32 bits of IPv6 multicast address are used in MAC address

<i>MAC multicast addr.</i>	<i>Used for</i>
01-00-5e-XX-XX-XX	IPv4 multicast
33-33-XX-XX-XX-XX	IPv6 multicast

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

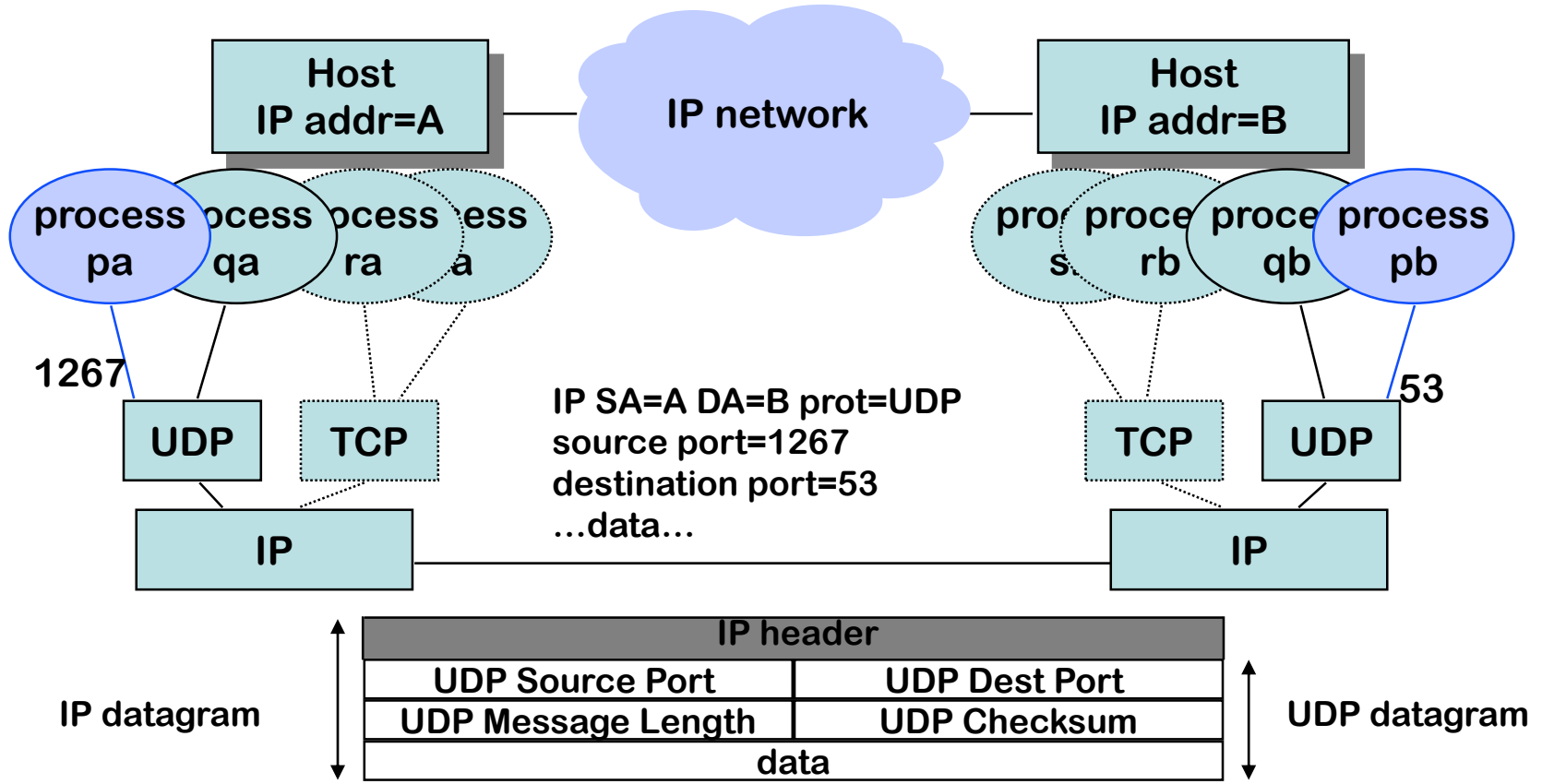
SLAAC Step 2: Duplicate Test

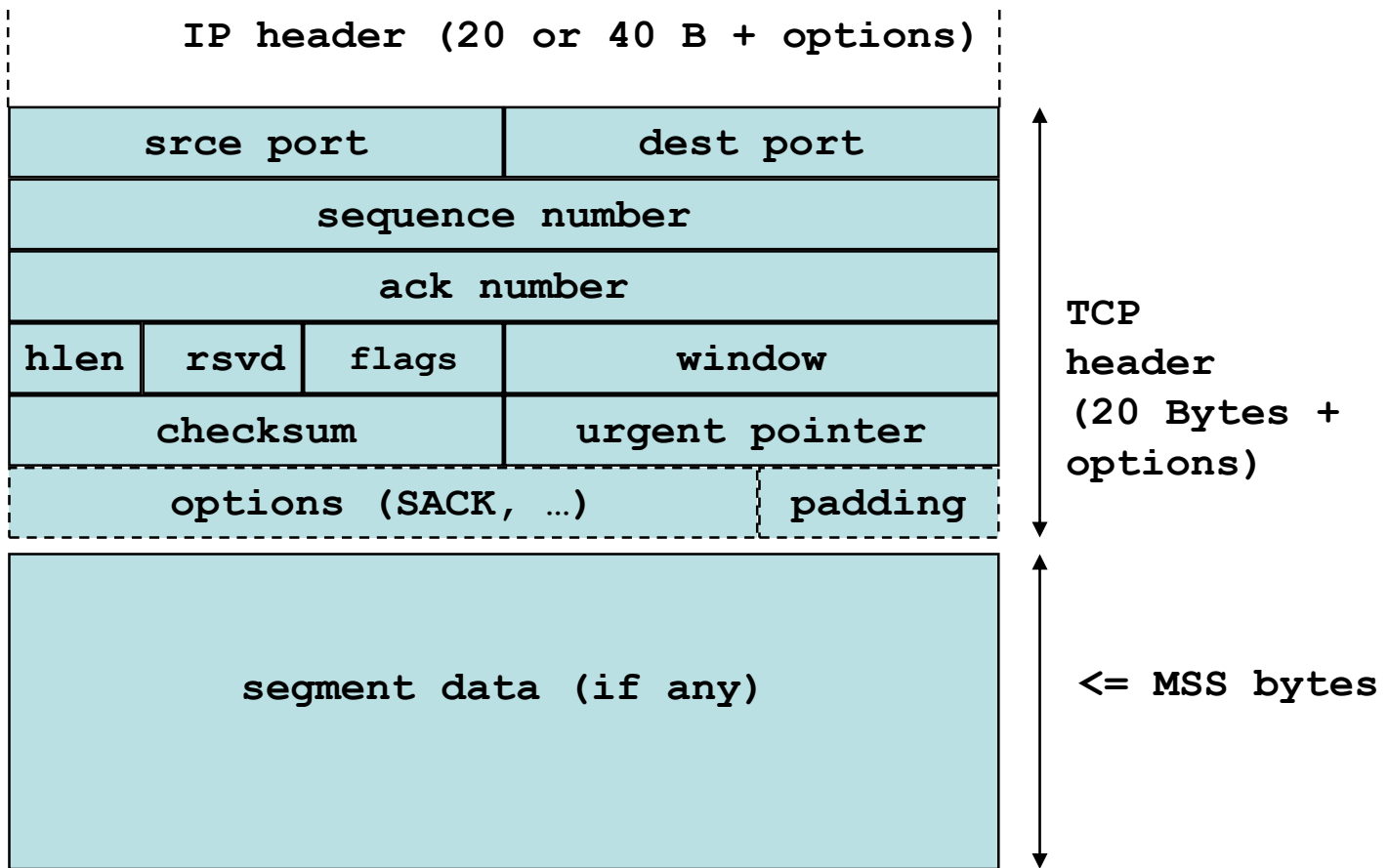


A sends a Neighbour Solicitation (NS) message to check for address duplication, sent to the **Solicited Node Multicast Address**.

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

UDP Uses Port Numbers





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 $0 \dots N$;
the algorithm
computes shortest
paths from node 0.
 $c(i,j)$: cost of link (i,j) .

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 .

```
 $m(0) = 0; m(i) = \infty \forall i \neq 0; V = \emptyset ; pred(i) = \emptyset \forall i;$   
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  $m(i) + c(i,j) = m(j)$   
                 $m(j) = m(i) + c(i,j)$   
                 $pred(j) = pred(j) \cup \{i\}$ 
```

Practical Aspects

OSPF packets 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 $k = 1, 2, \dots$ **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 ($\text{nextHop}(i)$)

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 $\text{nextHop}(i)$

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 $j == \text{nextHop}(i)$
 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, $\text{nextHop}(i) \leftarrow j$

The 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. A common sequence is

0. 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 (router-id of the BGP peer from whom route is received)

(The Cisco and FRR implementation of BGP, used in lab, have a few additional cases, not shown here)

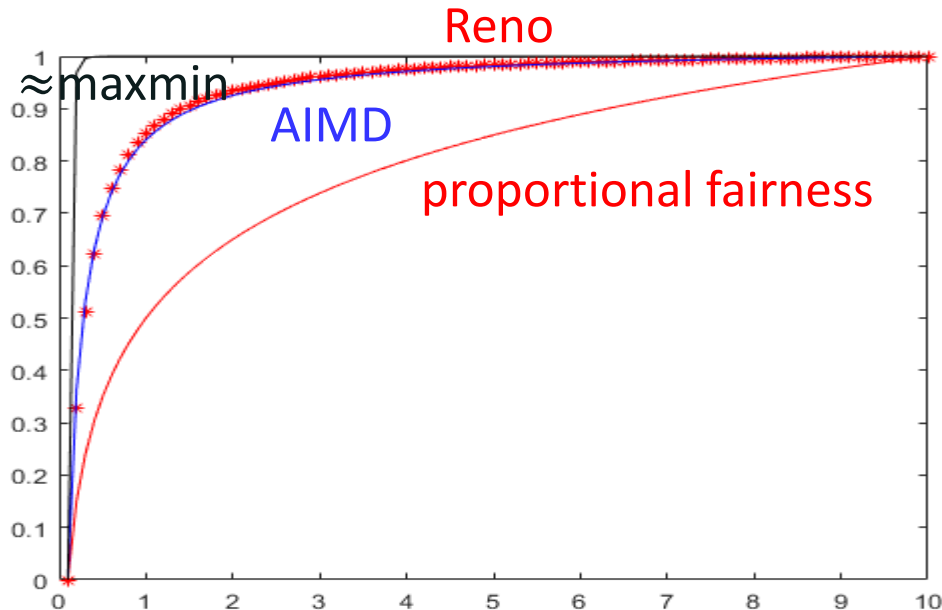
Fairness of TCP Reno

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 $\theta = \frac{MSS \cdot 1.22}{RTT \sqrt{q}}$

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

Cubic's Loss throughput formula

$$\theta \approx \max \left(\frac{1.054}{RTT^{0.25} q^{0.75}}, \frac{1.22}{RTT \sqrt{q}} \right)$$

in MSS per second.

Cubic's formula is same as Reno for small RTTs and small BW-delay products.

Other Cubic details

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

