

19.1

Data Communications and Networking **Fourth Edition**

Chapter 19 Network Layer: Logical Addressing

Copyright © The McGraw-Hill Companies, Inc. Permission required for reproduction or display.

An IPv4 address is a 32-bit address that uniquely and universally defines the connection of a device (for example, a computer or a router) to the Internet.

Topics discussed in this section:

Address Space Notations Classful Addressing Classless Addressing Network Address Translation (NAT)

An IPv4 address is 32 bits long.

The IPv4 addresses are unique and universal.

The address space of IPv4 is 2 ³² or 4,294,967,296.

Figure 19.1 *Dotted-decimal notation and binary notation for an IPv4 address*

Numbering systems are reviewed in Appendix B.

Change the following IPv4 addresses from binary notation to dotted-decimal notation.

- a. 10000001 00001011 00001011 11101111
- **b.** 11000001 10000011 00011011 11111111

Solution

We replace each group of 8 bits with its equivalent decimal number (see Appendix B) and add dots for separation.

- a. 129.11.11.239
- b. 193.131.27.255

Change the following IPv4 addresses from dotted-decimal notation to binary notation.

- a. 111.56.45.78
- b. $221.34.7.82$

Solution

We replace each decimal number with its binary equivalent (see Appendix B).

a. 01101111 00111000 00101101 01001110

b. 11011101 00100010 00000111 01010010

Example 19.3

Find the error, if any, in the following IPv4 addresses. a. 111.56.045.78

- b. 221.34.7.8.20
- c. $75.45.301.14$
- d. $11100010.23.14.67$

Solution

- *a. There must be no leading zero (045).*
- *b. There can be no more than four numbers.*
- *c. Each number needs to be less than or equal to 255.*
- *d. A mixture of binary notation and dotted-decimal notation is not allowed.*

In classful addressing, the address space is divided into five classes: A, B, C, D, and E.

Figure 19.2 *Finding the classes in binary and dotted-decimal notation*

a. Binary notation

b. Dotted-decimal notation

Find the class of each address.

- *a.* **0**0000001 00001011 00001011 11101111
- *b.* **110**00001 10000011 00011011 11111111
- *c.* **14**.23.120.8
- *d.* **252**.5.15.111

Solution

- *a. The first bit is 0. This is a class A address.*
- *b. The first 2 bits are 1; the third bit is 0. This is a class C address.*
- *c. The first byte is 14; the class is A.*
- *d. The first byte is 252; the class is E.*

Table 19.1 *Number of blocks and block size in classful IPv4 addressing*

In classful addressing, a large part of the available addresses were wasted.

Table 19.2 *Default masks for classful addressing*

Classful addressing, which is almost obsolete, is replaced with classless addressing.

Figure 19.3 shows a block of addresses, in both binary and dotted-decimal notation, granted to a small business that needs 16 addresses.

We can see that the restrictions are applied to this block. The addresses are contiguous. The number of addresses is a power of 2 (16 = 2⁴), and the first address is divisible by 16. The first address, when converted to a decimal number, is 3,440,387,360, which when divided by 16 results in 215,024,210.

Figure 19.3 *A block of 16 addresses granted to a small organization*

In IPv4 addressing, a block of addresses can be defined as x.y.z.t /*n* **in which x.y.z.t defines one of the addresses and the /***n* **defines the mask.**

The first address in the block can be found by setting the rightmost 32 − *n* **bits to 0s.**

A block of addresses is granted to a small organization. We know that one of the addresses is 205.16.37.39/28. What is the first address in the block?

Solution

The binary representation of the given address is 11001101 00010000 00100101 00100111 If we set 32−28 rightmost bits to 0, we get 11001101 00010000 00100101 0010000 or 205.16.37.32.

This is actually the block shown in Figure 19.3.

19.22

The last address in the block can be found by setting the rightmost 32 − n bits to 1s.

Find the last address for the block in Example 19.6.

Solution

The binary representation of the given address is 11001101 00010000 00100101 00100111 If we set 32 − 28 rightmost bits to 1, we get 11001101 00010000 00100101 00101111

or

205.16.37.47

This is actually the block shown in Figure 19.3.

The number of addresses in the block can be found by using the formula 2 32−n .

Find the number of addresses in Example 19.6.

Solution

The value of n is 28, which means that number of addresses is 2 32−28 or 16.

Another way to find the first address, the last address, and the number of addresses is to represent the mask as a 32 bit binary (or 8-digit hexadecimal) number. This is particularly useful when we are writing a program to find these pieces of information. In Example 19.5 the /28 can be represented as

11111111 11111111 11111111 11110000 (twenty-eight 1s and four 0s).

Find

19.27 *a. The first address b. The last address c. The number of addresses.*

Solution

a. The first address can be found by ANDing the given addresses with the mask. ANDing here is done bit by bit. The result of ANDing 2 bits is 1 if both bits are 1s; the result is 0 otherwise.

Example 19.9 (continued)

b. The last address can be found by ORing the given addresses with the complement of the mask. ORing here is done bit by bit. The result of ORing 2 bits is 0 if both bits are 0s; the result is 1 otherwise. The complement of a number is found by changing each 1 to 0 and each 0 to 1.

Example 19.9 (continued)

c. The number of addresses can be found by complementing the mask, interpreting it as a decimal number, and adding 1 to it.

0000000000 000000000 00000000 00001111 Mask complement: Number of addresses: $15 + 1 = 16$

Figure 19.4 *A network configuration for the block 205.16.37.32/28*

The first address in a block is normally not assigned to any device; it is used as the network address that represents the organization to the rest of the world.

Figure 19.5 *Two levels of hierarchy in an IPv4 address*

Figure 19.6 *A frame in a character-oriented protocol*

Each address in the block can be considered as a two-level hierarchical structure: the leftmost *n* **bits (prefix) define the network; the rightmost 32 − n bits define the host.**

Figure 19.7 *Configuration and addresses in a subnetted network*

19.36
Figure 19.8 *Three-level hierarchy in an IPv4 address*

An ISP is granted a block of addresses starting with 190.100.0.0/16 (65,536 addresses). The ISP needs to distribute these addresses to three groups of customers as follows:

- *a. The first group has 64 customers; each needs 256 addresses.*
- *b. The second group has 128 customers; each needs 128 addresses.*
- *c. The third group has 128 customers; each needs 64 addresses.*
- *Design the subblocks and find out how many addresses are still available after these allocations.*

Solution

Figure 19.9 shows the situation.

Group 1

For this group, each customer needs 256 addresses. This means that 8 (log2 256) bits are needed to define each host. The prefix length is then 32 − 8 = 24. The addresses

are

Group 2

For this group, each customer needs 128 addresses. This means that 7 (log2 128) bits are needed to define each host. The prefix length is then 32 − 7 = 25. The addresses are

Group 3

For this group, each customer needs 64 addresses. This means that 6 (log264) bits are needed to each host. The prefix length is then 32 − 6 = 26. The addresses are

Number of granted addresses to the ISP: 65,536 Number of allocated addresses by the ISP: 40,960 Number of available addresses: 24,576

Figure 19.9 *An example of address allocation and distribution by an ISP*

Table 19.3 *Addresses for private networks*

Figure 19.10 *A NAT implementation*

Site using private addresses

Figure 19.11 *Addresses in a NAT*

Figure 19.12 *NAT address translation*

Table 19.4 *Five-column translation table*

Figure 19.13 *An ISP and NAT*

Despite all short-term solutions, address depletion is still a long-term problem for the Internet. This and other problems in the IP protocol itself have been the motivation for IPv6.

Structure Address Space *Topics discussed in this section:*

An IPv6 address is 128 bits long.

Figure 19.14 *IPv6 address in binary and hexadecimal colon notation*

Figure 19.15 *Abbreviated IPv6 addresses*

Expand the address 0:15::1:12:1213 to its original.

Solution

We first need to align the left side of the double colon to the left of the original pattern and the right side of the double colon to the right of the original pattern to find how many 0s we need to replace the double colon.

> XXX:XXXX:XXXX:XXXX:XXXX:XXXX:XXXX $15:$ 1: 12:1213

This means that the original address is.

0000:0015:0000:0000:0000:0001:0012:1213

Table 19.5 *Type prefixes for IPv6 addresses*

Table 19.5 *Type prefixes for IPv6 addresses (continued)*

Figure 19.16 *Prefixes for provider-based unicast address*

Figure 19.17 *Multicast address in IPv6*

Figure 19.18 *Reserved addresses in IPv6*

Figure 19.19 *Local addresses in IPv6*

20.1

Data Communications and Networking **Fourth Edition**

Chapter 20 Network Layer: Internet Protocol

Copyright © The McGraw-Hill Companies, Inc. Permission required for reproduction or display.

In this section, we discuss internetworking, connecting networks together to make an internetwork or an internet.

Topics discussed in this section:

Need for Network Layer Internet as a Datagram Network Internet as a Connectionless Network

Figure 20.1 *Links between two hosts*

Figure 20.2 *Network layer in an internetwork*

Figure 20.3 *Network layer at the source, router, and destination*

a. Network layer at source

b. Network layer at destination

Figure 20.3 *Network layer at the source, router, and destination (continued)*

c. Network layer at a router

Switching at the network layer in the Internet uses the datagram approach to packet switching.

Communication at the network layer in the Internet is connectionless.

The Internet Protocol version 4 (IPv4) is the delivery mechanism used by the TCP/IP protocols.

Topics discussed in this section:

Datagram Fragmentation Checksum

Options

Figure 20.4 *Position of IPv4 in TCP/IP protocol suite*

Figure 20.5 *IPv4 datagram format*

Figure 20.6 *Service type or differentiated services*

The precedence subfield was part of version 4, but never used.
Table 20.1 *Types of service*

Table 20.2 *Default types of service*

Table 20.3 *Values for codepoints*

The total length field defines the total length of the datagram including the header.

Figure 20.7 *Encapsulation of a small datagram in an Ethernet frame*

Figure 20.8 *Protocol field and encapsulated data*

Table 20.4 *Protocol values*

An IPv4 packet has arrived with the first 8 bits as shown: 01000010

The receiver discards the packet. Why?

Solution

There is an error in this packet. The 4 leftmost bits (0100) show the version, which is correct. The next 4 bits (0010) show an invalid header length $(2 \times 4 = 8)$ *. The minimum number of bytes in the header must be 20. The packet has been corrupted in transmission.*

In an IPv4 packet, the value of HLEN is 1000 in binary. How many bytes of options are being carried by this packet?

Solution

The HLEN value is 8, which means the total number of bytes in the header is 8 × 4, or 32 bytes. The first 20 bytes are the base header, the next 12 bytes are the options.

In an IPv4 packet, the value of HLEN is 5, and the value of the total length field is 0x0028. How many bytes of data are being carried by this packet?

Solution

The HLEN value is 5, which means the total number of bytes in the header is 5 × 4, or 20 bytes (no options). The total length is 40 bytes, which means the packet is carrying 20 bytes of data (40 − 20).

An IPv4 packet has arrived with the first few hexadecimal digits as shown.

0x45000028000100000102 . . .

How many hops can this packet travel before being dropped? The data belong to what upper-layer protocol?

Solution

To find the time-to-live field, we skip 8 bytes. The time-tolive field is the ninth byte, which is 01. This means the packet can travel only one hop. The protocol field is the next byte (02), which means that the upper-layer protocol is IGMP.

Figure 20.9 *Maximum transfer unit (MTU)*

Table 20.5 *MTUs for some networks*

Figure 20.10 *Flags used in fragmentation*

D: Do not fragment M: More fragments

Figure 20.11 *Fragmentation example*

Figure 20.12 *Detailed fragmentation example*

A packet has arrived with an M bit value of 0. Is this the first fragment, the last fragment, or a middle fragment? Do we know if the packet was fragmented?

Solution

If the M bit is 0, it means that there are no more *fragments; the fragment is the last one. However, we cannot say if the original packet was fragmented or not. A non-fragmented packet is considered the last fragment.*

A packet has arrived with an M bit value of 1. Is this the first fragment, the last fragment, or a middle fragment? Do we know if the packet was fragmented?

Solution

If the M bit is 1, it means that there is at least one more fragment. This fragment can be the first one or a middle one, but not the last one. We don't know if it is the first one or a middle one; we need more information (the value of the fragmentation offset).

A packet has arrived with an M bit value of 1 and a fragmentation offset value of 0. Is this the first fragment, the last fragment, or a middle fragment?

Solution

Because the M bit is 1, it is either the first fragment or a middle one. Because the offset value is 0, it is the first fragment.

A packet has arrived in which the offset value is 100. What is the number of the first byte? Do we know the number of the last byte?

Solution

To find the number of the first byte, we multiply the offset value by 8. This means that the first byte number is 800. We cannot determine the number of the last byte unless we know the length.

A packet has arrived in which the offset value is 100, the value of HLEN is 5, and the value of the total length field is 100. What are the numbers of the first byte and the last byte?

Solution

The first byte number is 100 × 8 = 800. The total length is 100 bytes, and the header length is 20 bytes (5 × 4), which means that there are 80 bytes in this datagram. If the first byte number is 800, the last byte number must be 879.

Figure 20.13 shows an example of a checksum calculation for an IPv4 header without options. The header is divided into 16-bit sections. All the sections are added and the sum is complemented. The result is inserted in the checksum field.

Figure 20.13 *Example of checksum calculation in IPv4*

Figure 20.14 *Taxonomy of options in IPv4*

The network layer protocol in the TCP/IP protocol suite is currently IPv4. Although IPv4 is well designed, data communication has evolved since the inception of IPv4 in the 1970s. IPv4 has some deficiencies that make it unsuitable for the fast-growing Internet.

Topics discussed in this section:

Advantages Packet Format Extension Headers

Figure 20.15 *IPv6 datagram header and payload*

Figure 20.16 *Format of an IPv6 datagram*

Table 20.6 *Next header codes for IPv6*

Table 20.7 *Priorities for congestion-controlled traffic*

Table 20.8 *Priorities for noncongestion-controlled traffic*

Table 20.9 *Comparison between IPv4 and IPv6 packet headers*

Comparison

- 1. The header length field is eliminated in IPv6 because the length of the header is fixed in this version.
- 2. The service type field is eliminated in IPv6. The priority and flow label fields together take over the function of the service type field.
- 3. The total length field is eliminated in IPv6 and replaced by the payload length field.
- 4. The identification, flag, and offset fields are eliminated from the base header in IPv6. They are included in the fragmentation extension header.
- 5. The TTL field is called hop limit in IPv6.
- 6. The protocol field is replaced by the next header field.
- 7. The header checksum is eliminated because the checksum is provided by upper-layer protocols; it is therefore not needed at this level.
- 8. The option fields in IPv4 are implemented as extension headers in IPv6.

Figure 20.17 *Extension header types*

Table 20.10 *Comparison between IPv4 options and IPv6 extension headers*

Comparison

- 1. The no-operation and end-of-option options in IPv4 are replaced by Pad1 and PadN options in IPv6.
- 2. The record route option is not implemented in IPv6 because it was not used.
- 3. The timestamp option is not implemented because it was not used.
- 4. The source route option is called the source route extension header in IPv6.
- 5. The fragmentation fields in the base header section of IPv4 have moved to the fragmentation extension header in IPv6.
- 6. The authentication extension header is new in IPv6.
- 7. The encrypted security payload extension header is new in IPv6.

20-4 TRANSITION FROM IPv4 TO IPv6

Because of the huge number of systems on the Internet, the transition from IPv4 to IPv6 cannot happen suddenly. It takes a considerable amount of time before every system in the Internet can move from IPv4 to IPv6. The transition must be smooth to prevent any problems between IPv4 and IPv6 systems.

Dual Stack Tunneling Header Translation *Topics discussed in this section:*

Figure 20.18 *Three transition strategies*

Figure 20.19 *Dual stack*

Figure 20.20 *Tunneling strategy*

Figure 20.21 *Header translation strategy*

Table 20.11 *Header translation*

Header Translation Procedure

- 1. The IPv6 mapped address is changed to an IPv4 address by extracting the rightmost 32 bits.
- 2. The value of the IPv6 priority field is discarded.
- 3. The type of service field in IPv4 is set to zero.
- 4. The checksum for IPv4 is calculated and inserted in the corresponding field.
- 5. The IPv6 flow label is ignored.
- 6. Compatible extension headers are converted to options and inserted in the IPv4 header. Some may have to be dropped.
- 7. The length of IPv4 header is calculated and inserted into the corresponding field.
- 8. The total length of the IPv4 packet is calculated and inserted in the corresponding field.

Data Communications and Networking **Fourth Edition**

Chapter 21 Network Layer: Address Mapping, Error Reporting, and Multicasting

Copyright © The McGraw-Hill Companies, Inc. Permission required for reproduction or display.

The delivery of a packet to a host or a router requires two levels of addressing: logical and physical. We need to be able to map a logical address to its corresponding physical address and vice versa. This can be done by using either static or dynamic mapping.

Topics discussed in this section:

Mapping Logical to Physical Address Mapping Physical to Logical Address

Figure 21.1 *ARP operation*

a. ARP request is broadcast

b. ARP reply is unicast

Figure 21.2 *ARP packet*

Figure 21.3 *Encapsulation of ARP packet*

Figure 21.4 *Four cases using ARP*

Case 1. A host has a packet to send to another host on the same network.

Case 2. A host wants to send a packet to another host on another network. It must first be delivered to a router.

Case 3. A router receives a packet to be sent to a host on another network. It must first be delivered to the appropriate router.

Case 4. A router receives a packet to be sent to a host on the same network.

An ARP request is broadcast; an ARP reply is unicast.

A host with IP address 130.23.43.20 and physical address B2:34:55:10:22:10 has a packet to send to another host with IP address 130.23.43.25 and physical address A4:6E:F4:59:83:AB. The two hosts are on the same Ethernet network. Show the ARP request and reply packets encapsulated in Ethernet frames.

Solution

Figure 21.5 shows the ARP request and reply packets. Note that the ARP data field in this case is 28 bytes, and that the individual addresses do not fit in the 4-byte boundary. That is why we do not show the regular 4-byte boundaries for these addresses.

Figure 21.5 *Example 21.1, an ARP request and reply*

Figure 21.6 *Proxy ARP*

Figure 21.7 *BOOTP client and server on the same and different networks*

a. Client and server on the same network

b. Client and server on different networks

DHCP provides static and dynamic address allocation that can be manual or automatic.

21-2 ICMP

The IP protocol has no error-reporting or errorcorrecting mechanism. The IP protocol also lacks a mechanism for host and management queries. The Internet Control Message Protocol (ICMP) has been designed to compensate for the above two deficiencies. It is a companion to the IP protocol.

Topics discussed in this section:

Types of Messages Message Format Error Reporting and Query Debugging Tools

Figure 21.8 *General format of ICMP messages*

ICMP always reports error messages to the original source.

Figure 21.9 *Error-reporting messages*

Note

Important points about ICMP error messages: ❏ No ICMP error message will be generated in response to a datagram carrying an ICMP error message.

- **❏ No ICMP error message will be generated for a fragmented datagram that is not the first fragment.**
- **❏ No ICMP error message will be generated for a datagram having a multicast address.**
- **❏ No ICMP error message will be generated for a datagram having a special address such as 127.0.0.0 or 0.0.0.0.**

Figure 21.10 *Contents of data field for the error messages*

Figure 21.11 *Redirection concept*

Figure 21.12 *Query messages*

Figure 21.13 *Encapsulation of ICMP query messages*

Figure 21.14 shows an example of checksum calculation for a simple echo-request message. We randomly chose the identifier to be 1 and the sequence number to be 9. The message is divided into 16-bit (2-byte) words. The words are added and the sum is complemented. Now the sender can put this value in the checksum field.

Figure 21.14 *Example of checksum calculation*

We use the ping program to test the server fhda.edu. The result is shown on the next slide. The ping program sends messages with sequence numbers starting from 0. For each probe it gives us the RTT time. The TTL (time to live) field in the IP datagram that encapsulates an ICMP message has been set to 62. At the beginning, ping defines the number of data bytes as 56 and the total number of bytes as 84. It is obvious that if we add 8 bytes of ICMP header and 20 bytes of IP header to 56, the result is 84. However, note that in each probe ping defines the number of bytes as 64. This is the total number of bytes in the ICMP packet (56 + 8).

Example 21.3 (continued)

\$ ping fhda.edu

PING fhda.edu (153.18.8.1) 56 (84) bytes of data.

64 bytes from tiptoe.fhda.edu $(153.18.8.1)$: icmp_seq=0 64 bytes from tiptoe.fhda.edu $(153.18.8.1)$: icmp_seq=1 64 bytes from tiptoe.fhda.edu (153.18.8.1): icmp_seq=2 64 bytes from tiptoe.fhda.edu $(153.18.8.1)$: icmp_seq=3 64 bytes from tiptoe.fhda.edu (153.18.8.1): icmp_seq=4 64 bytes from tiptoe.fhda.edu $(153.18.8.1)$: icmp_seq=5 64 bytes from tiptoe.fhda.edu (153.18.8.1): icmp_seq=6 64 bytes from tiptoe.fhda.edu $(153.18.8.1)$: icmp_seq=7 64 bytes from tiptoe.fhda.edu $(153.18.8.1)$: icmp_seq=8 64 bytes from tiptoe.fhda.edu (153.18.8.1): icmp_seq=9 64 bytes from tiptoe.fhda.edu $(153.18.8.1)$: icmp_seq=10

--- fhda.edu ping statistics ---

11 packets transmitted, 11 received, 0% packet loss, time 10103ms rtt min/avg/max = $1.899/1.955/2.041$ ms

 $time=1.91$ ms

 $ttl = 62$

ttl= 62

tt $l=62$

ttl= 62

tt $l=62$

tt $l=62$

ttl= 62

tt $l=62$

- time= 2.04 ms
- $time=1.90$ ms
- $time=1.97$ ms
- $time=1.93$ ms
- $time=2.00$ ms
- $time=1.94$ ms
- $time=1.94$ ms
- ttl= 62 $time=1.97$ ms
- tt $l=62$ ttl= 62
	- $time=1.89$ ms
		- $time=1.98$ ms

Figure 21.15 *The traceroute program operation*

We use the traceroute program to find the route from the computer voyager.deanza.edu to the server fhda.edu. The following shows the result:

The unnumbered line after the command shows that the destination is 153.18.8.1. The packet contains 38 bytes: 20 bytes of IP header, 8 bytes of UDP header, and 10 bytes of application data. The application data are used by traceroute to keep track of the packets.

The first line shows the first router visited. The router is named Dcore.fhda.edu with IP address 153.18.31.254. The first round-trip time was 0.995 ms, the second was 0.899 ms, and the third was 0.878 ms. The second line shows the second router visited. The router is named Dbackup.fhda.edu with IP address 153.18.251.4. The three round-trip times are also shown. The third line shows the destination host. We know that this is the destination host because there are no more lines. The destination host is the server fhda.edu, but it is named tiptoe.fhda.edu with the IP address 153.18.8.1. The three round-trip times are also shown.

In this example, we trace a longer route, the route to *xerox.com (see next slide). Here there are 17 hops between source and destination. Note that some roundtrip times look unusual. It could be that a router was too busy to process the packet immediately.*

21-3 IGMP

The IP protocol can be involved in two types of communication: unicasting and multicasting. The Internet Group Management Protocol (IGMP) is one of the necessary, but not sufficient, protocols that is involved in multicasting. IGMP is a companion to the IP protocol.

Topics discussed in this section:

Group Management IGMP Messages and IGMP Operation Encapsulation Netstat Utility

Figure 21.16 *IGMP message types*

Figure 21.17 *IGMP message format*

Table 21.1 *IGMP type field*

Figure 21.18 *IGMP operation*

In IGMP, a membership report is sent twice, one after the other.

The general query message does not define a particular group.

Imagine there are three hosts in a network, as shown in Figure 21.19. A query message was received at time 0; the random delay time (in tenths of seconds) for each group is shown next to the group address. Show the sequence of report messages.

Solution

The events occur in this sequence:

a. Time 12: The timer for 228.42.0.0 in host A expires, and a membership report is sent, which is received by the router and every host including host B which cancels its timer for 228.42.0.0.

Example 21.6 (continued)

- *b. Time 30: The timer for 225.14.0.0 in host A expires, and a membership report is sent which is received by the router and every host including host C which cancels its timer for 225.14.0.0.*
- *c. Time 50: The timer for 238.71.0.0 in host B expires, and a membership report is sent, which is received by the router and every host.*
- *d. Time 70: The timer for 230.43.0.0 in host C expires, and a membership report is sent, which is received by the router and every host including host A which cancels its timer for 230.43.0.0.*

Figure 21.19 *Example 21.6*

Figure 21.20 *Encapsulation of IGMP packet*

21.41

The IP packet that carries an IGMP packet has a value of 1 in its TTL field.

Table 21.2 *Destination IP addresses*

Figure 21.21 *Mapping class D to Ethernet physical address*

21.44

An Ethernet multicast physical address is in the range 01:00:5E:00:00:00 to 01:00:5E:7F:FF:FF.

Change the multicast IP address 230.43.14.7 to an Ethernet multicast physical address.

Solution

We can do this in two steps:

a. We write the rightmost 23 bits of the IP address in hexadecimal. This can be done by changing the rightmost 3 bytes to hexadecimal and then subtracting 8 from the leftmost digit if it is greater than or equal to 8. In our example, the result is 2B:0E:07.

Example 21.7 (continued)

b. We add the result of part a to the starting Ethernet multicast address, which is 01:00:5E:00:00:00. The result is

Change the multicast IP address 238.212.24.9 to an Ethernet multicast address.

Solution

- *a. The rightmost 3 bytes in hexadecimal is D4:18:09. We need to subtract 8 from the leftmost digit, resulting in 54:18:09.*
- *b. We add the result of part a to the Ethernet multicast starting address. The result is*

Figure 21.22 *Tunneling*

Unicast IP datagram

We use netstat (see next slide) with three options: -n, -r, and -a. The -n option gives the numeric versions of IP addresses, the -r option gives the routing table, and the -a option gives all addresses (unicast and multicast). Note that we show only the fields relative to our discussion. "Gateway" defines the router, "Iface" defines the interface.

Note that the multicast address is shown in color. Any packet with a multicast address from 224.0.0.0 to *239.255.255.255 is masked and delivered to the Ethernet interface.*

21.50

Example 21.9 (continued)

21-4 ICMPv6

We discussed IPv6 in Chapter 20. Another protocol that has been modified in version 6 of the TCP/IP protocol suite is ICMP (ICMPv6). This new version follows the same strategy and purposes of version 4.

Error Reporting Query *Topics discussed in this section:*

Figure 21.23 *Comparison of network layers in version 4 and version 6*

Table 21.3 *Comparison of error-reporting messages in ICMPv4 and ICMPv6*

Table 21.4 *Comparison of query messages in ICMPv4 and ICMPv6*

22.1

Data Communications and Networking **Fourth Edition**

Chapter 22 Network Layer: Delivery, Forwarding, and Routing

Copyright © The McGraw-Hill Companies, Inc. Permission required for reproduction or display.

The network layer supervises the handling of the packets by the underlying physical networks. We define this handling as the delivery of a packet.

Direct Versus Indirect Delivery *Topics discussed in this section:*

Figure 22.1 *Direct and indirect delivery*

22.3

Forwarding means to place the packet in its route to its destination. Forwarding requires a host or a router to have a routing table. When a host has a packet to send or when a router has received a packet to be forwarded, it looks at this table to find the route to the final destination.

Topics discussed in this section:

Forwarding Techniques Forwarding Process Routing Table

Figure 22.2 *Route method versus next-hop method*

22.5

Figure 22.3 *Host-specific versus network-specific method*

Figure 22.4 *Default method*

Figure 22.5 *Simplified forwarding module in classless address*

In classless addressing, we need at least four columns in a routing table.

Make a routing table for router R1, using the configuration in Figure 22.6.

Solution Table 22.1 shows the corresponding table.

Figure 22.6 *Configuration for Example 22.1*

22.11

Table 22.1 *Routing table for router R1 in Figure 22.6*

Show the forwarding process if a packet arrives at R1 in Figure 22.6 with the destination address 180.70.65.140. Solution

- *The router performs the following steps:*
- *1. The first mask (/26) is applied to the destination address. The result is 180.70.65.128, which does not match the corresponding network address.*
- *2. The second mask (/25) is applied to the destination address. The result is 180.70.65.128, which matches the corresponding network address. The next-hop address and the interface number m0 are passed to ARP for further processing.*

Show the forwarding process if a packet arrives at R1 in Figure 22.6 with the destination address 201.4.22.35.

Solution

The router performs the following steps:

- *1. The first mask (/26) is applied to the destination address. The result is 201.4.22.0, which does not match the corresponding network address.*
- *2. The second mask (/25) is applied to the destination address. The result is 201.4.22.0, which does not match the corresponding network address (row 2).*
3. The third mask (/24) is applied to the destination address. The result is 201.4.22.0, which matches the corresponding network address. The destination address of the packet and the interface number m3 are passed to ARP.

Show the forwarding process if a packet arrives at R1 in Figure 22.6 with the destination address 18.24.32.78.

Solution

This time all masks are applied, one by one, to the destination address, but no matching network address is found. When it reaches the end of the table, the module gives the next-hop address 180.70.65.200 and interface number m2 to ARP. This is probably an outgoing package that needs to be sent, via the default router, to someplace else in the Internet.

Figure 22.7 *Address aggregation*

Routing table for R1

Figure 22.8 *Longest mask matching*

Routing table for R3

As an example of hierarchical routing, let us consider Figure 22.9. A regional ISP is granted 16,384 addresses starting from 120.14.64.0. The regional ISP has decided to divide this block into four subblocks, each with 4096 addresses. Three of these subblocks are assigned to three local ISPs; the second subblock is reserved for future use. Note that the mask for each block is /20 because the original block with mask /18 is divided into 4 blocks.

The first local ISP has divided its assigned subblock into 8 smaller blocks and assigned each to a small ISP. Each small ISP provides services to 128 households, each using four addresses.

The second local ISP has divided its block into 4 blocks and has assigned the addresses to four large organizations.

The third local ISP has divided its block into 16 blocks and assigned each block to a small organization. Each small organization has 256 addresses, and the mask is / 24.

There is a sense of hierarchy in this configuration. All routers in the Internet send a packet with destination address 120.14.64.0 to 120.14.127.255 to the regional ISP.

Figure 22.9 *Hierarchical routing with ISPs*

Figure 22.10 *Common fields in a routing table*

One utility that can be used to find the contents of a routing table for a host or router is netstat in UNIX or LINUX. The next slide shows the list of the contents of a default server. We have used two options, r and n. The option r indicates that we are interested in the routing table, and the option n indicates that we are looking for numeric addresses. Note that this is a routing table for a host, not a router. Although we discussed the routing table for a router throughout the chapter, a host also needs a routing table.

The destination column here defines the network address. The term gateway used by UNIX is synonymous with router. This column actually defines the address of the next hop. The value 0.0.0.0 shows that the delivery is direct. The last entry has a flag of G, which means that the destination can be reached through a router (default router). The Iface defines the interface.

More information about the IP address and physical address of the server can be found by using the ifconfig command on the given interface (eth0).

\$ ifconfig eth0 eth0 Link encap: Ethernet HWaddr 00: B0: D0: DF: 09: 5D inet addr:153.18.17.11 Bcast:153.18.31.255 Mask:255.255.240.0

Figure 22.11 *Configuration of the server for Example 22.6*

22-3 UNICAST ROUTING PROTOCOLS

A routing table can be either static or dynamic. A static table is one with manual entries. A dynamic table is one that is updated automatically when there is a change somewhere in the Internet. A routing protocol is a combination of rules and procedures that lets routers in the Internet inform each other of changes.

Topics discussed in this section:

Optimization Intra- and Interdomain Routing Distance Vector Routing and RIP Link State Routing and OSPF Path Vector Routing and BGP

Figure 22.12 *Autonomous systems*

Figure 22.13 *Popular routing protocols*

Figure 22.14 *Distance vector routing tables*

Figure 22.15 *Initialization of tables in distance vector routing*

In distance vector routing, each node shares its routing table with its immediate neighbors periodically and when there is a change.

Figure 22.16 *Updating in distance vector routing*

Figure 22.17 *Two-node instability*

Figure 22.18 *Three-node instability*

Figure 22.19 *Example of a domain using RIP*

Figure 22.20 *Concept of link state routing*

Figure 22.21 *Link state knowledge*

Figure 22.22 *Dijkstra algorithm*

Figure 22.23 *Example of formation of shortest path tree*

Table 22.2 *Routing table for node A*

Figure 22.24 *Areas in an autonomous system*

Figure 22.25 *Types of links*

Figure 22.26 *Point-to-point link*

Figure 22.27 *Transient link*

a. Transient network

b. Unrealistic representation

c. Realistic representation

Figure 22.28 *Stub link*

a. Stub network

b. Representation

Figure 22.29 *Example of an AS and its graphical representation in OSPF*

a. Autonomous system

b. Graphical representation

Figure 22.30 *Initial routing tables in path vector routing*

Figure 22.31 *Stabilized tables for three autonomous systems*

Figure 22.32 *Internal and external BGP sessions*

22-4 MULTICAST ROUTING PROTOCOLS

In this section, we discuss multicasting and multicast routing protocols.

Unicast, Multicast, and Broadcast Applications Multicast Routing Routing Protocols *Topics discussed in this section:*

Figure 22.33 *Unicasting*

In unicasting, the router forwards the received packet through only one of its interfaces.

Figure 22.34 *Multicasting*

In multicasting, the router may forward the received packet through several of its interfaces.

Figure 22.35 *Multicasting versus multiple unicasting*

a. Multicasting

b. Multiple unicasting

Emulation of multicasting through multiple unicasting is not efficient and may create long delays, particularly with a large group.

In unicast routing, each router in the domain has a table that defines a shortest path tree to possible destinations.

Figure 22.36 *Shortest path tree in unicast routing*

In multicast routing, each involved router needs to construct a shortest path tree for each group.

Figure 22.37 *Source-based tree approach*

In the source-based tree approach, each router needs to have one shortest path tree for each group.

Figure 22.38 *Group-shared tree approach*

In the group-shared tree approach, only the core router, which has a shortest path tree for each group, is involved in multicasting.

Figure 22.39 *Taxonomy of common multicast protocols*

Multicast link state routing uses the source-based tree approach.

Flooding broadcasts packets, but creates loops in the systems.

RPF eliminates the loop in the flooding process.

Figure 22.40 *Reverse path forwarding (RPF)*

Figure 22.41 *Problem with RPF*

Figure 22.42 *RPF Versus RPB*

a. RPF

b. RPB

RPB creates a shortest path broadcast tree from the source to each destination. It guarantees that each destination receives one and only one copy of the packet.

Figure 22.43 *RPF, RPB, and RPM*

c. RPM (after pruning)

d. RPM (after grafting)

RPM adds pruning and grafting to RPB to create a multicast shortest path tree that supports dynamic membership changes.

Figure 22.44 *Group-shared tree with rendezvous router*

Shared tree

Figure 22.45 *Sending a multicast packet to the rendezvous router*

In CBT, the source sends the multicast packet (encapsulated in a unicast packet) to the core router. The core router decapsulates the packet and forwards it to all interested interfaces.

PIM-DM is used in a dense multicast environment, such as a LAN.

PIM-DM uses RPF and pruning and grafting strategies to handle multicasting. However, it is independent of the underlying unicast protocol.

PIM-SM is used in a sparse multicast environment such as a WAN.

PIM-SM is similar to CBT but uses a simpler procedure.

Figure 22.46 *Logical tunneling*

Figure 22.47 *MBONE*

