

Eric Wang/Xuhui Wang CSC361 P2 documentation

ARP packets:

286	9.087706000	2a:f4:c0:ad:27:1c	8e:9a:e1:90:64:8b	ARP	42	Who has 10.0.0.2? Tell 10.0.0.1
287	9.087849000	8e:9a:e1:90:64:8b	2a:f4:c0:ad:27:1c	ARP	42	Who has 10.0.0.1? Tell 10.0.0.2
288	9.087855000	2a:f4:c0:ad:27:1c	8e:9a:e1:90:64:8b	ARP	42	10.0.0.1 is at 2a:f4:c0:ad:27:1c
289	9.087884000	8e:9a:e1:90:64:8b	2a:f4:c0:ad:27:1c	ARP	42	10.0.0.2 is at 8e:9a:e1:90:64:8b

UDP packets:

Filter:		▼	Expression...	Clear	Apply	Save
No.	Time	Source	Destination	Protocol	Length	Info
82	4.086157000	10.0.0.2	10.0.0.1	UDP	76	Source port: 59363 Destination port: 59363
83	4.086174000	10.0.0.1	10.0.0.2	UDP	76	Source port: http Destination port: http
84	4.086219000	10.0.0.2	10.0.0.1	UDP	76	Source port: 59363 Destination port: 59363
85	4.086236000	10.0.0.1	10.0.0.2	UDP	76	Source port: http Destination port: http
86	4.086280000	10.0.0.2	10.0.0.1	UDP	76	Source port: 59363 Destination port: 59363
87	4.086297000	10.0.0.1	10.0.0.2	UDP	49	Source port: http Destination port: http
88	4.086323000	10.0.0.2	10.0.0.1	UDP	76	Source port: 59363 Destination port: 59363
89	4.086339000	10.0.0.1	10.0.0.2	UDP	49	Source port: http Destination port: http
90	4.086364000	10.0.0.2	10.0.0.1	UDP	76	Source port: 59363 Destination port: 59363
91	4.086379000	10.0.0.1	10.0.0.2	UDP	49	Source port: http Destination port: http
92	4.086404000	10.0.0.2	10.0.0.1	UDP	76	Source port: 59363 Destination port: 59363
93	4.086420000	10.0.0.1	10.0.0.2	UDP	76	Source port: http Destination port: http
94	4.086464000	10.0.0.2	10.0.0.1	UDP	76	Source port: 59363 Destination port: 59363
95	4.086482000	10.0.0.1	10.0.0.2	UDP	76	Source port: http Destination port: http
96	4.086526000	10.0.0.2	10.0.0.1	UDP	76	Source port: 59363 Destination port: 59363
97	4.086543000	10.0.0.1	10.0.0.2	UDP	76	Source port: http Destination port: http
98	4.086588000	10.0.0.2	10.0.0.1	UDP	76	Source port: 59363 Destination port: 59363
99	4.086605000	10.0.0.1	10.0.0.2	UDP	49	Source port: http Destination port: http
100	4.086631000	10.0.0.2	10.0.0.1	UDP	76	Source port: 59363 Destination port: 59363
101	4.086647000	10.0.0.1	10.0.0.2	UDP	76	Source port: http Destination port: http
102	4.086663000	10.0.0.2	10.0.0.1	UDP	76	Source port: 59363 Destination port: 59363
Data (34 bytes)						
000	2a f4 c0 ad 27 1c 8e 9a e1 90 64 8b 08 00 45 00	*...'....d...E.				
010	00 3e c5 30 40 00 40 11 61 7c 0a 00 00 02 0a 00	.>.0@.@. a				
020	00 01 e7 e3 00 50 00 2a 14 3e 70 69 6e 67 20 33P.* .>ping 3				
030	31 20 32 30 32 30 2d 30 33 2d 30 34 20 31 34 3a	1 2020-03-04 14:				

Client terminal:

```
host: h2
PING 77 2020-03-04 14:16:18.475722
the RRT is 0.000116 s.
PING 78 2020-03-04 14:16:18.475864
the RRT is 3.5e-05 s.
PING 79 2020-03-04 14:16:18.475926
the RRT is 3.6e-05 s.
PING 80 2020-03-04 14:16:18.475988
the RRT is 3.5e-05 s.
PING 81 2020-03-04 14:16:18.476050
the RRT is 3.5e-05 s.
PING82 message has dropped and needs to be retransmitted.
PING 82 2020-03-04 14:16:18.476112
the RRT is 7.6e-05 s.
PING 83 2020-03-04 14:16:18.476215
the RRT is 3.5e-05 s.
PING 84 2020-03-04 14:16:18.476276
the RRT is 3.5e-05 s.
PING 85 2020-03-04 14:16:18.476338
the RRT is 3.5e-05 s.
PING 86 2020-03-04 14:16:18.476399
the RRT is 3.6e-05 s.
PING 87 2020-03-04 14:16:18.476461
the RRT is 3.5e-05 s.
PING88 message has dropped and needs to be retransmitted.
PING 88 2020-03-04 14:16:18.476522
the RRT is 7.6e-05 s.
PING89 message has dropped and needs to be retransmitted.
started Firefox in a while. Do you want to clean it up for a fresh, like-new experience? And
```

Retransmission:

```
# Assign IP address and port number to socket

serverSocket.bind((sys.argv[1], int(sys.argv[2])))
print("The server is ready to receive!")

while True:
    # Receive the client packet along with the address it is coming from
    msg, clientAddress = serverSocket.recvfrom(2048)

    rand = random.randint(1, 10)

    if rand < 4:
        reMsg = "dropped"
        serverSocket.sendto(reMsg.encode(), clientAddress)
        continue

    modifiedMessage = msg.decode().upper()
    serverSocket.sendto(modifiedMessage.encode(), clientAddress)
```

I implemented retransmission on the server side as indicated in the code above:

If the server decides to drop the packet, it sends a string "dropped" to indicate the client that the packet last sent by the client was dropped.

```
dropped = False
msg = "ping " + str(i) + " " + datetime.datetime.now().strftime("%Y-%m-%d %H:%M:%S.%f")
start = datetime.datetime.now()
clientSocket.sendto(msg.encode(), (serverName, serverPort))
modifiedMsg, serverAddress = clientSocket.recvfrom(2048)

while modifiedMsg.decode() == "dropped":
    dropped = True
    print("PING" + str(i) + " message has dropped and needs to be retransmitted.")
    print()
    clientSocket.sendto(msg.encode(), (serverName, serverPort))
    modifiedMsg, serverAddress = clientSocket.recvfrom(2048)

end = datetime.datetime.now()
print(modifiedMsg.decode())
print("the RRT is " + str((end - start).total_seconds()) + " s.")
print()
#time.sleep(0.2)
```

I implemented retransmission on the client side as indicated in the code above:

If the client receives the packet replied by the server containing the string "dropped", it resends the same packet to the server until it gets the right replied packet.

RRT:

```
msg = "ping " + str(i) + " " + datetime.datetime.now().strftime("%Y-%m-%d %H:%M:%S.%T")
start = datetime.datetime.now()
clientSocket.sendto(msg.encode(), (serverName, serverPort))
modifiedMsg, serverAddress = clientSocket.recvfrom(2048)

while modifiedMsg.decode() == "dropped":
    dropped = True
    print("PING" + str(i) + " message has dropped and needs to be retransmitted.")
    print()
    clientSocket.sendto(msg.encode(), (serverName, serverPort))
    modifiedMsg, serverAddress = clientSocket.recvfrom(2048)

end = datetime.datetime.now()
print(modifiedMsg.decode())
print("the RRT is " + str((end - start).total_seconds()) + " s.")
print()
```

I implemented the RRT calculation as indicated in the code above:

Before I send the packet to the server, I get the start timestamp.

I get the end timestamp after I have received the replied packet correctly (including the retransmission time if need).

Then I get the time difference between start and end, which is the RRT.