1. What are the first and last packets for the POST request?

From packet **4** which has sequence number **1** to packet **199** which has sequence number **164041**.

We can find information about these segments in Wireshark in the packet-detail pane.

```
to segment duta (so bytes)
[122 Reassembled TCP Segments (164090 bytes): #4(565), #5
    [Frame: 4, payload: 0-564 (565 bytes)]
     [Frame: 5, payload: 565-2024 (1460 bytes)]
    [Frame: 7, payload: 2025-3484 (1460 bytes) segment
     [Frame: 8, payload: 3485-4944 (1460 bytes)]
     [Frame: 10, payload: 4945-6404 (1460 bytes)]
     [Frame: 11, payload: 6405-7864 (1460 bytes)]
     [Frame: 13, payload: 7865-9011 (1147 bytes)]
     [Frame: 18, payload: 9012-10471 (1460 bytes)]
     [Frame: 19. pavload: 10472-11931 (1460 bytes)]
 [Frame: 194, payload: 159388-160847 (1460 bytes)]
 [Frame: 195, payload: 160848-162307 (1460 bytes)]
 [Frame: 196, payload: 162308-163767 (1460 bytes)]
 [Frame: 197, payload: 163768-164039 (272 bytes)]
[Frame: 199, payload: 164040-164089 (50 bytes)]
 [Segment count: 122]
                                Last segment
```

2. What is the IP address and the TCP port number used by the client computer (source) that is transferring the file to gaia.cs.umass.edu?

192.168.1.102 and port number is 1161. See information below!

```
▶ Internet Protocol Version 4, Src: 192.168.1.102, Dst: 128.119.245.12
▼ Transmission Control Protocol, Src Port: 1161, Dst Port: 80, Seq: 0, Len: 0
Source Port: 1161
```

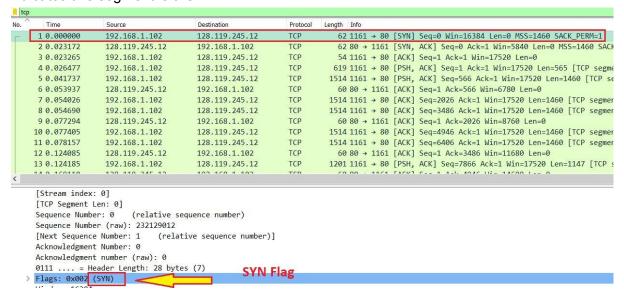
3. What is the IP address of gaia.cs.umass.edu? On what port number is it sending and receiving TCP segments for this connection?

128.119.245.12 and port 80 is used for this connection. See information below!

```
▶ Internet Protocol Version 4, Src: 128.119.245.12, Dst: 192.168.1.102
▼ Transmission Control Protocol, Src Port: 80, Dst Port: 1161, Seq: 0, Ack: 1, Len: 0
Source Port: 80
```

4. What is the sequence number of the TCP SYN segment that is used to initiate the TCP connection between the client computer and gaia.cs.umass.edu? What is it in the segment that identifies the segment as a SYN segment?

Actual sequence number for TCP SYN is 232129012. It has a flag (SYN) that indicates this segment is a SYN.



5. What is the sequence number of the SYNACK segment sent by gaia.cs.umass.edu to the client computer in reply to the SYN? What is the value of the ACKnowledgement field in the SYNACK segment? How did gaia.cs.umass.edu determine that value? What is it in the segment that identifies the segment as a SYNACK segment?

Sequence number: 883061785

ACKnowledgment number: 232129013

This number is based on the last packet sequence number. In this case the last sequence number is **232129012** and therefore the gaia.cs.umass.edu send acknowledgement with number **232129013** which means that it has received data from to **232129012**.

The flag name (SYN, ACK) is displayed in the segment field.

Time	Source	Destination	Protocol	Length Info
1 0.000000	192.168.1.102	128.119.245.12	TCP	62 1161 → 80 [SYN] Seg=0 Win=16384 Len=0 MSS=1460 SACK_
2 0.023172	128.119.245.12	192.168.1.102	TCP	62 80 → 1161 [SYN, ACK] Seq=0 Ack=1 Win=5840 Len=0 NSS=
3 0.023265	192.168.1.102	128.119.245.12	TCP	54 1161 → 80 [ACK] Seq=1 Ack=1 Win=17520 Len=0
4 0.026477	192.168.1.102	128.119.245.12	TCP	619 1161 → 80 [PSH, ACK] Seq=1 Ack=1 Win=17520 Len=565 [
5 0.041737	192.168.1.102	128.119.245.12	TCP	1514 1161 → 80 [PSH, ACK] Seq=566 Ack=1 Win=17520 Len=146
6 0.053937	128.119.245.12	192.168.1.102	TCP	60 80 → 1161 [ACK] Seq=1 Ack=566 Win=6780 Len=0
7 0.054026	192.168.1.102	128.119.245.12	TCP	1514 1161 → 80 [ACK] Seq=2026 Ack=1 Win=17520 Len=1460 [T
8 0.054690	192.168.1.102	128.119.245.12	TCP	1514 1161 → 80 [ACK] Seq=3486 Ack=1 Win=17520 Len=1460 [T
9 0.077294	128.119.245.12	192.168.1.102	TCP	60 80 → 1161 [ACK] Seq=1 Ack=2026 Win=8760 Len=0
10 0.077405	192.168.1.102	128.119.245.12	TCP	1514 1161 → 80 [ACK] Seq=4946 Ack=1 Win=17520 Len=1460 [T
11 0.078157	192.168.1.102	128.119.245.12	TCP	1514 1161 → 80 [ACK] Seq=6406 Ack=1 Win=17520 Len=1460 [T
12 0.124085	128.119.245.12	192.168.1.102	TCP	60 80 → 1161 [ACK] Seq=1 Ack=3486 Win=11680 Len=0
13 0.124185	192.168.1.102	128.119.245.12	TCP	1201 1161 → 80 [PSH, ACK] Seq=7866 Ack=1 Win=17520 Len=11
44.0.400440	400 440 045 40	100 100 1 100	TCD	CO DO - 4404 FACKT C 4 A-I- 4040 HI- 44000 H 0
[Next Sequence Acknowledgment Acknowledgment	Len: 0] er: 0 (relative so er (raw): 883061785 e Number: 1 (rela t Number: 1 (rela t number (raw): 2321			
> Flags: 0x012	eader Length: 28 byte	es (/)		

6. What is the sequence number of the TCP segment containing the HTTP POST command?

Packet 4: **232129013**. See picture below!

	1 0.000000	192.168.1.102	128.119.245.1
	2 0.023172	128.119.245.12	192.168.1.102
	3 0.023265	192.168.1.102	128.119.245.1
	4 0.026477	192.168.1.102	128.119.245.1
Ī	5 0.041737	192.168.1.102	128.119.245.1
	6 0.053937	128.119.245.12	192.168.1.102
	7 0.054026	192.168.1.102	128.119.245.1
	8 0.054690	192.168.1.102	128.119.245.1
	9 0.077294	128.119.245.12	192.168.1.102
	10 0.077405	192.168.1.102	128.119.245.1
	11 0.078157	192.168.1.102	128.119.245.1
	12 0.124085	128.119.245.12	192.168.1.102
	13 0.124185	192.168.1.102	128.119.245.1
4			

```
▶ Frame 4: 619 bytes on wire (4952 bits), 619 bytes captured (4952 bits), 619 bytes captur
```

▶ Internet Protocol Version 4, Src: 192.168.1.102, Dst: 128.119.24
▼ Transmission Control Protocol, Src Port: 1161, Dst Port: 80, Sec

```
Source Port: 1161
Destination Port: 80
[Stream index: 0]
[TCP Segment Len: 565]
Sequence number: 1 (relative sequence number)
Sequence number (raw): 232129013
[Next sequence number: 566 (relative sequence number)]
Acknowledgment number: 1 (relative ack number)
Acknowledgment number (raw): 883061786
```

7. Consider the TCP segment containing the HTTP POST as the first segment in the TCP connection. What are the sequence numbers of the first six segments in the TCP connection (including the segment containing the HTTP POST)? At what time was each segment sent? When was the ACK for each segment received? Given the difference between when each TCP segment was sent, and when its acknowledgement was received, what is the RTT value for each of the six segments? What is the EstimatedRTT value (see Section 3.5.3, page 269 in text) after the receipt of each ACK? Assume that the value of the EstimatedRTT is equal to the measured RTT for the first segment, and then is computed using the EstimatedRTT equation on page 270 for all subsequent segments.

0101 = Header Length: 20 bytes (5)

Note: Wireshark has a nice feature that allows you to plot the RTT for each of the TCP segments sent. Select a TCP segment in the "listing of captured packets" window that is being sent from the client to the gaia.cs.umass.edu server. Then select: Statistics->TCP Stream Graph->Round Trip Time Graph.

0	4 . 232129013	sent time:	0.026477	Ack time: 0.053937 (6)
0	5 . 232129578	sent time:	0.041737	Ack time: 0.077294 (9)
0	7 . 883061786	sent time:	0.054026	Ack time: 0.124085 (12)
0	8. 232131038	sent time:	0.054690	Ack time: 0 .169118 (14)
0	10 . 232132498	sent time:	0.077405	Ack time: 0.217299 (15)
0	11 . 883061786	sent time:	0.078157	Ack time: 0.267802 (16)

Here is an example of how we calculated RTT for each segment. Note that this information can be found in Wireshark also.

RTT value for segment 4 = 0.053937 - 0.026477 = **0.02746**

RTT values

4 = 0.02746

5 = 0.03555

7 = 0.07005

8 = 0.11442

10 = 0.13989

11 = 0.18964

EstimatedRTT = 0.875 * EstimatedRTT + 0.125 * SampleRTT(RTT value)

An example how we calculate Estimated RTT for packet 5:

Estimated RTT(5)= 0.875 (constant) * 0.0274 (prev. EstRTT) + 0.125 (constant) * 0.03555(RTT value)

Estimated RTT

4 = 0.0274

5 = 0.0285

7 = 0.0337

8 = 0.0437

10 = 0.0558

11 = 0.0725

- 8. What is the length of each of the first six TCP segments?
 - **4**. 565
 - **5**. 1460
 - **7**. 1460
 - **8.** 1460
 - **10**. 1460

11. 1460

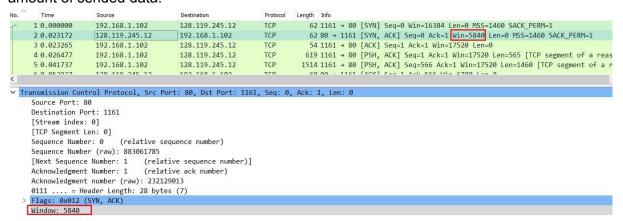
This information is taken from the packet-listing pane in Wireshark but also can be found in packet-detail pane for each segment as well.

4 0.026477	192.168.1.102	128.119.245.12	TCP	619 1161 → 80 [PSH, ACK] Seq=1 Ack=1 Win=17520 Let=565 TCP segment of a
5 0.041737	192.168.1.102	128.119.245.12	TCP	1514 1161 → 80 [PSH, ACK] Seq=566 Ack=1 Win=17520 Len-1460 [TCP segment c
7 0.054026	192.168.1.102	128.119.245.12	TCP	1514 1161 → 80 [ACK] Seq=2026 Ack=1 Win=17520 Len 1460 [TCP segment of a
8 0.054690	192.168.1.102	128.119.245.12	TCP	1514 1161 → 80 [ACK] Seq=3486 Ack=1 Win=17520 Len 1460 TCP segment of a
10 0.077405	192.168.1.102	128.119.245.12	TCP	1514 1161 → 80 [ACK] Seq=4946 Ack=1 Win=17520 Len€1460 TCP segment of a
11 0.078157	192.168.1.102	128.119.245.12	TCP	1514 1161 → 80 [ACK] Seq=6406 Ack=1 Win=17520 Len 1460 TCP segment of a

9. What is the minimum amount of available buffer space advertised at the receiver for the entire trace? Does the lack of receiver buffer space ever throttle the sender?

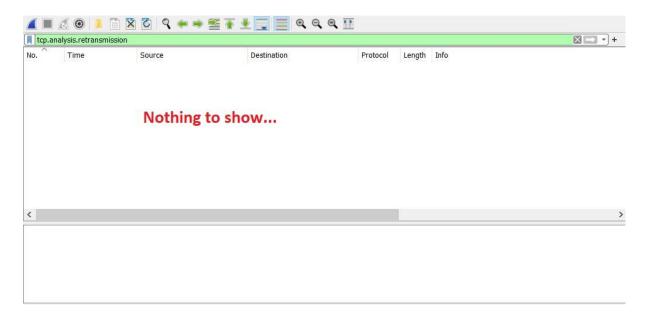
Minimum amount buffer space is 5840 bytes which has sequence number 883061785

The lack of receiver buffer space never throttles the sender because the available size of buffer at server increases after each ACK sent. Because the maximum amount of bytes in each segment is 1460 there is no limit for sender to decrease the amount of sended data.



10. Are there any retransmitted segments in the trace file? What did you check for (in the trace) in order to answer this question?

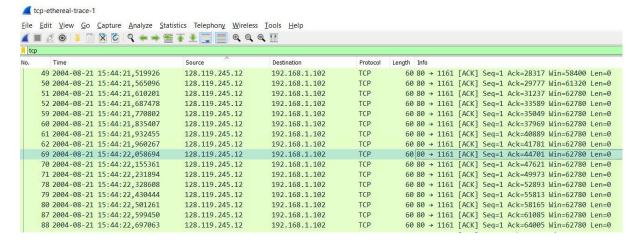
No, we used following filter which shows retransmitted segments in Wireshark: *tcp.analysis.retransmission*



11. How much data does the receiver typically acknowledge in an ACK? Can you identify cases where the receiver is ACKing every other received segment (see Table 3.2 on page 278 in the text).

Receiver typically acknowledge 1460 bytes in an ACK, which after observing the ACKs sent by receiver we discovered that receiver is ACKing every 1460 bytes.

Yes, for example packet No 69 is an Ack that acknowledged two segments (segment No 63 and 64). We see that the previous segment is Ack for 41781 but the next one is 44701 which include two segments of length 1460.



12. What is the throughput (bytes transferred per unit time) for the TCP connection? Explain how you calculated this value.

Throughput = amount of data transmitted / time incurred

Throughput = 164090 / (5,297341 - 0,026477) = 31136.62 B/s

We found amount of data transmitted in packet-detail pane (see following screenshot)

```
[Frame: 185, payload: 155576-156467 (892 bytes)]
[Frame: 192, payload: 156468-157927 (1460 bytes)]
[Frame: 193, payload: 157928-159387 (1460 bytes)]
[Frame: 194, payload: 159388-160847 (1460 bytes)]
[Frame: 195, payload: 160848-162307 (1460 bytes)]
[Frame: 196, payload: 162308-163767 (1460 bytes)]
[Frame: 197, payload: 163768-164039 (272 bytes)]
[Frame: 199, payload: 164040-164089 (50 bytes)]
[Segment count: 122]
[Reassembled TCP length: 164090]
[Reassembled TCP Data: 504+5354202f657468657265616c2d6c6162732f6c6162332d312d7265706c792e68746d...]
```

By dividing the amount of data transmitted by the time passed for transmission we calculated the throughput. Note that we calculated the time from which the first TCP segment after triple-handshake was sent which is packet No 4.

TASK A:

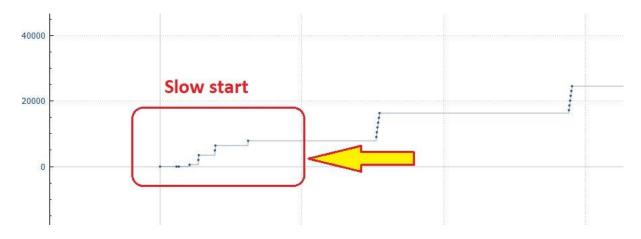
According to our observations from the questions, we got to understand that there is a three-way handshake needed in order to establish a connection between client and the server. This way, the client sends a SYN packet including it's sequence number asking the server for permission to establish the connection. Then the server responds to the client with a SYNACK packet including a sequence number which means that it confirms the request. At last, the client ACKs the server's response and starts sending the segments. The next steps will be taken recursively, which means that for every packet sent by client, the server ACKs the packets and determines the next packet by telling it's sequence number.

EstimatedRTT is the estimated round trip time that client determines itself to know the time it takes for the server to ACK the packets. In case, there is no signal from the server in the EstimatedRTT, then the client assumes that the packet is lost and will opt to send the packet again, this is how clients interpret packet losses.

TASK B:

13. Use the *Time-Sequence-Graph (Stevens)* plotting tool to view the sequence number versus time plot of segments being sent from the client to the gaia.cs.umass.edu server. Can you identify if and where TCP's *slow start* phase begins and ends, as well as if and where *congestion avoidance* takes over? Comment on ways in which the measured data differs from the idealized behavior of TCP that we've studied in the text.

After our observations, we could identify that the TCP's slow start begins at packet 1 to packet 13. The sender is starting to send a small amount of segments in each time and after every good ack (ack in time) it increases the congestion window size until it reaches an ideal size. And according to the graph below, we guess that congestion avoidance phase takes over after packet 14.



Slow start and congestion avoidance are **mandatory** components of TCP, differing in how they increase the size of **cwnd** in response to received ACKs.(Computer Networking - A Top-down Approach, page 266)

14. Explain the relationship between (i) the congestion window (cwnd), (ii) the receiver advertised window (rwnd), (iii) the number of unacknowledged bytes, and (iv) the effective window at the sender (i.e., the window effectively limiting the data transmission).

congestion window(cwnd): is an additional variable in TCP congestion-control mechanism which determines the amount of unacknowledged data at a sender.

Suppose Host **A** is sending Host **B** a large file over a TCP connection. The number of *unacknowledged bytes* that **A** sends cannot exceed the size of the receive buffer.

receiver advertised window(rwnd): is a variable that provides the flow-control service. It is typically the free spare space in the receiving buffer. How? Host **B** tells Host **A** how much spare room it has in the connection buffer by placing its current value of *rwnd* in the receive window field of every segment it sends to **A**.

effective window: The size of the recent advertised window depends on each ACK sent by the server to the client.

LastByteSent - LastByteAcked <= min{cwnd, rwnd}

15. Is it generally possible to find the congestion window size (cwnd) and how it changes with time, from the captured trace files? If so, please explain how. If not, please explain when and when not. Motivate your answer and give examples.

According **RFC5681**, CONGESTION WINDOW (cwnd): A TCP state variable that limits the amount of data a TCP can send. At any given time, a TCP MUST NOT send data with a sequence number higher than the sum of the highest acknowledged sequence number and the minimum of cwnd and rwnd.

So it is not generally possible to find the *cwnd* size as it is not advertised, However it is possible to find the *rwnd* size, because it is advertised.

TASK C:

16. What is the throughput of each of the connections in bps (bits per second)? What is the total bandwidth of the host on which the clients are running? Discuss the TCP fairness for this case.

Connection	Total transferred bytes	Duration (in seconds)	RTT (in milliseconds)
1	165095720	521	12
2	165842766	521	12
3	165458792	514	12
4	163235772	512	12

Connection1: 165095720 / 521 * 8 = 2535059,04 b/s

Connection2: 165842766 / 521 * 8 = 2546529,99 b/s

Connection3: 165458792 / 514 *8 = 2575234.12 b/s

Connection4: 163235772 / 512 * 8 = 2550558.94 b/s

Total bandwidth = 10207382,09 bits

In this case, we have almost similar durations and identical RTTs.

By looking at the throughput in each connection we do not see very large differences so we assume that it is a fair TCP connection.

17. What is the throughput of each of the connections in bps (bits per second)? What is the total bandwidth of the host on which the clients are running? Discuss the TCP fairness for this case

Connection	Total transferred bytes	Duration (in seconds)	RTT (in milliseconds)
1	261319130	90	13
2	175995832	90	35
3	151894552	90	68
4	140388568	90	73
5	108610702	90	49
6	70644690	90	33
7	65744938	90	135
8	43212876	90	326
9	39222524	90	322

Connection 1: 261319130 / 90 * 8 = 23,228,367.1111 b/s

Connection 2: 175995832 / 90 * 8 = 15,644,073.9556 b/s

Connection 3: 151894552 / 90 * 8 = 13,501,737.9556 b/s

Connection 4: 140388568 / 90 * 8 = 12,478,983.8222 b/s

Connection 5: 108610702 / 90 * 8 = 9,654,284.62222 b/s

Connection 6: 70644690 / 90 * 8 = 6,279,528 b/s

Connection 7: 65744938 / 90 *8 = 5,843,994.4889 b/s

Connection 8: 43212876 / 90 * 8 = 3,841,144.5334 b/s

Connection 9: 39222524 / 90 * 8 = 3,486,446.5778 b/s

Total bandwidth is: 93958561.06682

In this case, we have almost identical durations but different RTTs.

On contrary to the previous question, we see a very large difference in throughput between these 9 connections which means that TCP is not fair in this case and bandwidth has been unfairly shared between these 9 clients (running in the same host).

18. Discuss the TCP fairness for this case. How does it differ from the previous cases, and how is it affected by the use of BitTorrent?

Connection	Total transferred bytes	Duration (in seconds)	RTT (in milliseconds)
1	108851134	58	40
2	90435681	58	36
3	57971584	53	100
4	32000012	29	68
5	32557334	35	31
6	27199361	31	33
7	26329578	31	122
8	38834490	56	146
9	23571761	35	74
10	36252962	55	66

Connection 1: 108851134 / 58 * 8 = 15013949.5172 b/s

Connection 2: 90435681 / 58 * 8 = 12473887.0345 b/s

Connection 3: 57971584 / 53 *8 = 8750427.77358 b/s

Connection 4: 32000012 / 29 * 8 = 8827589.51724 b/s

Connection 5: 32557334 / 35 * 8 = 7441676.34286 b/s

Connection 6: 27199361 / 31 * 8 = 7019189.93548 b/s

Connection 7: 26329578 / 31 *8 = 6794729.80645 b/s

Connection 8: 38834490 / 56 * 8 = 5547784.28571 b/s

Connection 9: 23571761 / 35 * 8 = 5387831.08571 b/s

Connection 10: 36252962 / 55 * 8 = 5273158.10909 b/s

In this case, unlike the previous two questions, we have different durations and RTTs. However when we compare connections throughput with each other, we see that they are more likely to be **fair** rather than **unfair**. For example, connections **3 and 4**, **5 and 6**, **8**, **9 and 10**, have respectively very similar throughputs.

These clients benefit from using BitTorrent and help each other with downloading the same file from the server. Each client may share those parts of data it downloaded to another client and it causes a more fair connection.