The various stages a TCP connection can be in (setting up, tearing down, transmitting data, etc.) are complicated



The various stages a TCP connection can be in (setting up, tearing down, transmitting data, etc.) are complicated

There is a standard TCP state diagram that describes how TCP should act in most cases

The various stages a TCP connection can be in (setting up, tearing down, transmitting data, etc.) are complicated

There is a standard TCP state diagram that describes how TCP should act in most cases

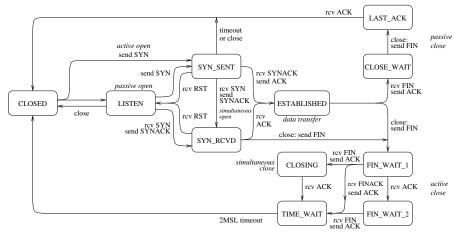
Though it only covers non-error cases: it does not say what to do if, say, a SYNFIN segment arrives

The various stages a TCP connection can be in (setting up, tearing down, transmitting data, etc.) are complicated

There is a standard TCP state diagram that describes how TCP should act in most cases

Though it only covers non-error cases: it does not say what to do if, say, a SYNFIN segment arrives

And it shows little about timeouts and retransmissions



TCP State Diagram

We start (and end) in CLOSED



There are the two opens: active and passive



There are the two opens: active and passive

LISTEN is a server waiting for a connection



There are the two opens: active and passive

LISTEN is a server waiting for a connection

ESTABLISHED is the normal data transfer state



There are the two opens: active and passive

LISTEN is a server waiting for a connection

ESTABLISHED is the normal data transfer state

And the two closes: active and passive



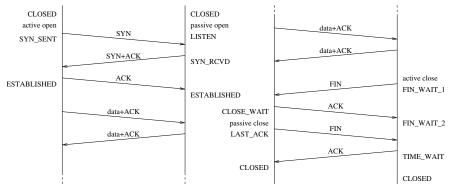
There are the two opens: active and passive

LISTEN is a server waiting for a connection

ESTABLISHED is the normal data transfer state

And the two closes: active and passive

This state diagram is followed for each end of a connection, i.e., each socket in the socketpair

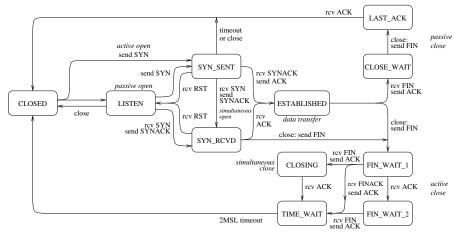


Typical TCP Timeline

The active close is somewhat complicated by the need for reliability

The active close is somewhat complicated by the need for reliability

The TIME_WAIT state (also called 2MSL state) appears before the final close: the active-close end of the connection must remain non-closed until a time period has passed



TCP State Diagram





In a perfect world this would be enough to close the connection



In a perfect world this would be enough to close the connection

But we have to deal with the case of the final ACK being lost



In a perfect world this would be enough to close the connection

But we have to deal with the case of the final ACK being lost

And resent if it didn't get to the other end





The *maximum segment lifetime* (MSL) is a value that represents the longest time a segment can live in the network before being discarded (probably through TTL expiry)



The *maximum segment lifetime* (MSL) is a value that represents the longest time a segment can live in the network before being discarded (probably through TTL expiry)

This was originally defined to be 2 minutes, but implementations often choose smaller values, like 60 seconds



The *maximum segment lifetime* (MSL) is a value that represents the longest time a segment can live in the network before being discarded (probably through TTL expiry)

This was originally defined to be 2 minutes, but implementations often choose smaller values, like 60 seconds

A TCP connection is required stay in TIME_WAIT for twice the MSL

This is in case the final ACK (of the final FIN) was lost and needs to be retransmitted

This is in case the final ACK (of the final FIN) was lost and needs to be retransmitted

The OS has to keep the connection hanging around for a little to cover this case

This is in case the final ACK (of the final FIN) was lost and needs to be retransmitted

The OS has to keep the connection hanging around for a little to cover this case

Even if the process that used the connection has exited

This is in case the final ACK (of the final FIN) was lost and needs to be retransmitted

The OS has to keep the connection hanging around for a little to cover this case

Even if the process that used the connection has exited

And while in this wait state if a new process tries to make a connection using the same ports it will be denied: the old connection is still active. We don't want to deliver late packets to the new process

This is in case the final ACK (of the final FIN) was lost and needs to be retransmitted

The OS has to keep the connection hanging around for a little to cover this case

Even if the process that used the connection has exited

And while in this wait state if a new process tries to make a connection using the same ports it will be denied: the old connection is still active. We don't want to deliver late packets to the new process

In this sense the TCP connection and the process using it are quite separate entities

When an application exits, the OS sends FINs on behalf of the application for all currently open connections. This makes sure everything is tidied up nicely (even if the programmer didn't)

When an application exits, the OS sends FINs on behalf of the application for all currently open connections. This makes sure everything is tidied up nicely (even if the programmer didn't)

And if it was an active close, OS needs to hold the connection in the 2MSL state for a while: the connection definitely outlives the application!

When an application exits, the OS sends FINs on behalf of the application for all currently open connections. This makes sure everything is tidied up nicely (even if the programmer didn't)

And if it was an active close, OS needs to hold the connection in the 2MSL state for a while: the connection definitely outlives the application!

If a host is shut down normally, rather than crashing, the operating system will (should!) send FINs for all currently open connections

When an application exits, the OS sends FINs on behalf of the application for all currently open connections. This makes sure everything is tidied up nicely (even if the programmer didn't)

And if it was an active close, OS needs to hold the connection in the 2MSL state for a while: the connection definitely outlives the application!

If a host is shut down normally, rather than crashing, the operating system will (should!) send FINs for all currently open connections

It really should do the TIME_WAIT, but often implementations don't bother as this would hold up the shutdown

TCP Strategies

We now take a look at how TCP manages to get the best out of a connection

TCP Strategies

We now take a look at how TCP manages to get the best out of a connection

For example: TCP gets reliability by acknowledging every byte sent. Does this mean two segments for every data packet: one data packet out, one ACK packet back?

TCP Strategies

We now take a look at how TCP manages to get the best out of a connection

For example: TCP gets reliability by acknowledging every byte sent. Does this mean two segments for every data packet: one data packet out, one ACK packet back?

It is possible to implement TCP like this, but performance would be poor

TCP Strategies

We now take a look at how TCP manages to get the best out of a connection

For example: TCP gets reliability by acknowledging every byte sent. Does this mean two segments for every data packet: one data packet out, one ACK packet back?

It is possible to implement TCP like this, but performance would be poor

So a typical TCP implementation will be a bit more smart on its use of ACKs: we have already mentioned delaying an ACK to let it piggyback on a returning data segment



That is just first of many strategies a TCP implementation can employ while still following the TCP protocol



That is just first of many strategies a TCP implementation can employ while still following the TCP protocol

We shall look at a few basic strategies, starting with more detail on the advertised window



As data arrives at its destination the OS puts it into a buffer, ready for the receiving application to read it. We have already seen the TCP *advertised window* in a returning segment which indicates how much of this buffer space is left



As data arrives at its destination the OS puts it into a buffer, ready for the receiving application to read it. We have already seen the TCP *advertised window* in a returning segment which indicates how much of this buffer space is left

The space left depends on

- how fast the sender is sending the data
- how fast the application is reading the data



As data arrives at its destination the OS puts it into a buffer, ready for the receiving application to read it. We have already seen the TCP *advertised window* in a returning segment which indicates how much of this buffer space is left

The space left depends on

- how fast the sender is sending the data
- how fast the application is reading the data

If the data arrives faster than it is read, the buffer will fill up





It is a *sliding window* mechanism, used as a form of flow control



It is a *sliding window* mechanism, used as a form of flow control

Imagine the bytes being sent as a long stream, starting at byte 0 (actually byte *n*, given by the initial sequence number) and going up



It is a *sliding window* mechanism, used as a form of flow control

Imagine the bytes being sent as a long stream, starting at byte 0 (actually byte n, given by the initial sequence number) and going up

A sliding window describes the range of bytes in the stream the sender can transmit next

Advertised Window

As the window gets smaller, the sender should send more slowly



Advertised Window

As the window gets smaller, the sender should send more slowly

As the window gets bigger, the sender can send more quickly

As the window gets smaller, the sender should send more slowly

As the window gets bigger, the sender can send more quickly

The sender recomputes the space available in the receiver every time it receives an ACK

Advertised Window

The left hand edge of the window is defined by the acknowledgement number in the latest ACK

Advertised Window

The left hand edge of the window is defined by the acknowledgement number in the latest ACK

The right hand edge is then given by adding on the size of the advertised window

Advertised Window

The left hand edge of the window is defined by the acknowledgement number in the latest ACK

The right hand edge is then given by adding on the size of the advertised window

The window size is sent in every ACK segment

Advertised Window

The left hand edge of the window is defined by the acknowledgement number in the latest ACK

The right hand edge is then given by adding on the size of the advertised window

The window size is sent in every ACK segment

As more ACKs are received, the window *closes* as the left edge advances

Advertised Window

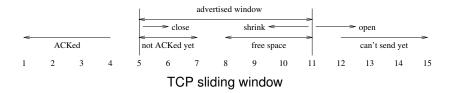
As the application reads data, the window *opens* as the right edge advances

Advertised Window

As the application reads data, the window *opens* as the right edge advances

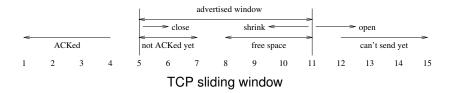
Rarely, the window can *shrink* (right edge recedes), perhaps if the buffer shrinks due to the memory being needed elsewhere

Advertised Window



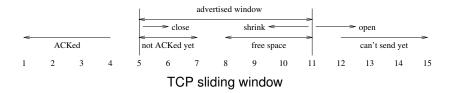
This is from the point of view of the sending end of a connection;

Advertised Window



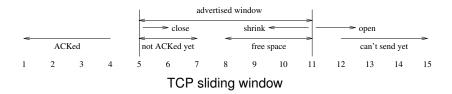
The situation is that we have just sent a segment with bytes 5-7; then received an ACK of 5 with a window of 7;

Advertised Window



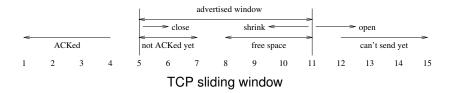
Bytes to the left of the window (1-4) have been ACKed and are safe in the destination;

Advertised Window



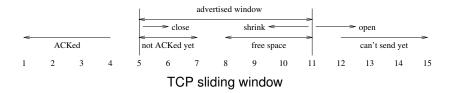
The advertised window tells us there is space for 7 bytes in the destination: bytes to the right (12 onwards) cannot be sent yet as the destination has nowhere to put them;

Advertised Window



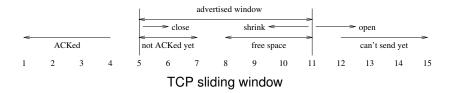
Bytes within the window are either not ACKed yet, or represent free space;

Advertised Window



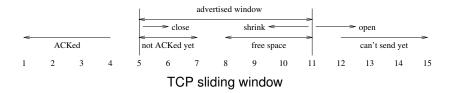
unACKed bytes (5-7) are those that have been sent by the sender, possibly received by the destination, and an ACK not yet received by the sender and possibly not yet sent by the receiver;

Advertised Window



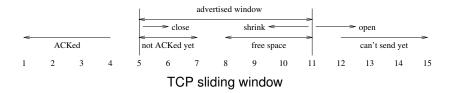
The free space (8-11) is the actual number of bytes that the sender can be sure that can be buffered;

Advertised Window



The sender can compute this free space as the latest window value minus the number of bytes sent but as yet unACKed;

Advertised Window



Thus the sender knows the limit on how much more data it can currently send





The sender will have to wait before sending more data



The sender will have to wait before sending more data

When the receiver is ready to receive more data it will send a duplicate ACK with the same ACK number as the ACK with window 0, but now with a non-zero window: this is a *window update segment*



The sender will have to wait before sending more data

When the receiver is ready to receive more data it will send a duplicate ACK with the same ACK number as the ACK with window 0, but now with a non-zero window: this is a *window update segment*

It may or may not contain data itself



The sender will have to wait before sending more data

When the receiver is ready to receive more data it will send a duplicate ACK with the same ACK number as the ACK with window 0, but now with a non-zero window: this is a *window update segment*

It may or may not contain data itself

Complications arise if this window update gets lost: the *Persist Timer* (see later) is used here



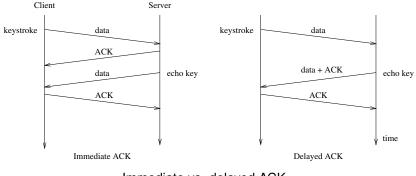
The next strategy we have mentioned before



The next strategy we have mentioned before

Instead of immediately ACKing every segment, we can slightly delay it and *piggyback* it on returning data

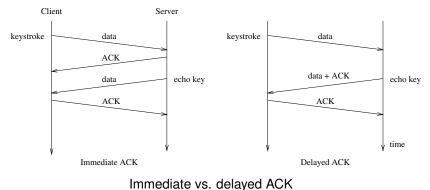
TCP Strategies Delayed ACKs



Immediate vs. delayed ACK

For example, when logged in to a remote terminal each keystroke is echoed back to your screen;

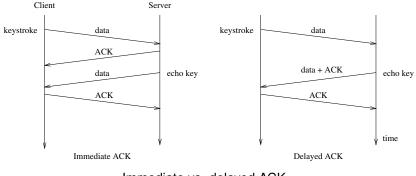
TCP Strategies Delayed ACKs



.

An immediate ACK would use four segments;

TCP Strategies Delayed ACKs



Immediate vs. delayed ACK

A delayed ACK piggybacking on the data for the echoed key uses just three segments



As far as the user is concerned, they see the keystroke echo in the same way, with no extra delay, but fewer segments are sent



As far as the user is concerned, they see the keystroke echo in the same way, with no extra delay, but fewer segments are sent

It is important to reduce the traffic on a heavily loaded network



As far as the user is concerned, they see the keystroke echo in the same way, with no extra delay, but fewer segments are sent

It is important to reduce the traffic on a heavily loaded network

It also reduces the chance of a lost segment

TCP Strategies Delayed ACKs

By delaying, we might also be able to ACK more than one segment at a time



If we receive, say, two segments in a period we are delaying, we can simply ACK the last segment: this implicitly ACKs the previous two segments



If we receive, say, two segments in a period we are delaying, we can simply ACK the last segment: this implicitly ACKs the previous two segments

An ACK is actually about acknowledging bytes, not acknowledging segments, but will usually align with segments



If we receive, say, two segments in a period we are delaying, we can simply ACK the last segment: this implicitly ACKs the previous two segments

An ACK is actually about acknowledging bytes, not acknowledging segments, but will usually align with segments

So an ACK indicates which byte we are expecting next and says all previous bytes have been safely received



If we receive, say, two segments in a period we are delaying, we can simply ACK the last segment: this implicitly ACKs the previous two segments

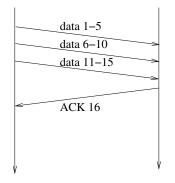
An ACK is actually about acknowledging bytes, not acknowledging segments, but will usually align with segments

So an ACK indicates which byte we are expecting next and says all previous bytes have been safely received

This reduces traffic again

TCP Strategies

Delayed ACKs



ACKing bytes received

ACKs acknowledge bytes received, not segments

TCP Strategies Delayed ACKs

So how long to delay an ACK?



So how long to delay an ACK?

If too long, the sender might think the segment was lost and resend



So how long to delay an ACK?

If too long, the sender might think the segment was lost and resend

If too short, we do not get so many free piggybacks or multiple ACKed segments



So how long to delay an ACK?

If too long, the sender might think the segment was lost and resend

If too short, we do not get so many free piggybacks or multiple ACKed segments

A typical implementation will delay for up to 200ms



The TCP specification says you should send an ACK for at least every second full-sized segment and you *must not* delay for more than 500ms



The TCP specification says you should send an ACK for at least every second full-sized segment and you *must not* delay for more than 500ms

This one of the many timers associated with TCP



The TCP specification says you should send an ACK for at least every second full-sized segment and you *must not* delay for more than 500ms

This one of the many timers associated with TCP

Each time you receive a data segment the TCP software should set a timer for that segment that expires after 200ms





Many operating systems have a single global timer that fires every 200ms rather than a timer per segment received



Many operating systems have a single global timer that fires every 200ms rather than a timer per segment received

When the timer goes off, all unACKed segments are ACKed



Many operating systems have a single global timer that fires every 200ms rather than a timer per segment received

When the timer goes off, all unACKed segments are ACKed

Not so accurate as per-segment timers, but much easier to implement





If you get an out-of-order segment (its sequence number is not the one you are expecting next, e.g., a segment was possibly lost), you *must not* delay, but send an ACK immediately



If you get an out-of-order segment (its sequence number is not the one you are expecting next, e.g., a segment was possibly lost), you *must not* delay, but send an ACK immediately

This might well be a duplicate ACK of one you sent earlier. This is to inform the sender as soon as possible that something might have gone wrong



If you get an out-of-order segment (its sequence number is not the one you are expecting next, e.g., a segment was possibly lost), you *must not* delay, but send an ACK immediately

This might well be a duplicate ACK of one you sent earlier. This is to inform the sender as soon as possible that something might have gone wrong

Though the other end will wait for three duplicate ACKs just to be sure before resending





A keystroke could be 1 byte



A keystroke could be 1 byte

This is sent in a TCP segment that has 20 bytes of header



- A keystroke could be 1 byte
- This is sent in a TCP segment that has 20 bytes of header
- This is contained in a IP datagram with 20 bytes of header



- A keystroke could be 1 byte
- This is sent in a TCP segment that has 20 bytes of header
- This is contained in a IP datagram with 20 bytes of header
- And so on down the layers





Such a packet is called a tinygram



Such a packet is called a *tinygram*

The proliferation of tinygrams causes additional congestion in a network



Such a packet is called a *tinygram*

The proliferation of tinygrams causes additional congestion in a network

Nagle created a strategy for reducing this



Such a packet is called a *tinygram*

The proliferation of tinygrams causes additional congestion in a network

Nagle created a strategy for reducing this

It applies to the sender of the tinygram (client)



Nagle's Algorithm:

a TCP connection can have only one outstanding un-ACKed small segment: no additional small segments can be sent until that ACK has been received



Nagle's Algorithm:

a TCP connection can have only one outstanding un-ACKed small segment: no additional small segments can be sent until that ACK has been received

If you are sending tinygrams, only send one and wait until you get its ACK before sending any more



Nagle's Algorithm:

a TCP connection can have only one outstanding un-ACKed small segment: no additional small segments can be sent until that ACK has been received

If you are sending tinygrams, only send one and wait until you get its ACK before sending any more

Any small segments waiting to be sent should be collected together into a single larger segment that is sent when the ACK is received



This segment can also be sent if either (a) you collect enough small segments to fill a MSS segment, or (b) they have collectively exceeded half the destination's advertised window size



This segment can also be sent if either (a) you collect enough small segments to fill a MSS segment, or (b) they have collectively exceeded half the destination's advertised window size

This leaves open the definition of "small"



This segment can also be sent if either (a) you collect enough small segments to fill a MSS segment, or (b) they have collectively exceeded half the destination's advertised window size

This leaves open the definition of "small"

Variants choose anything from "1 byte" to "any segment shorter than the maximum segment size"



Note that when window scaling is in effect, "small" must be at least the size of the window scale factor, as we can't advertise a window smaller than that



Note that when window scaling is in effect, "small" must be at least the size of the window scale factor, as we can't advertise a window smaller than that

But that won't be a constraint until the scale is bigger than a segment, e.g., $2^{10} = 1024$, but $2^{11} = 2048 > 1500$



This is a very simple strategy and reduces the number of tinygrams without introducing extra perceived delay (over that delay there is there already)



This is a very simple strategy and reduces the number of tinygrams without introducing extra perceived delay (over that delay there is there already)

The faster ACKs come back, the more tinygrams can be sent



This is a very simple strategy and reduces the number of tinygrams without introducing extra perceived delay (over that delay there is there already)

The faster ACKs come back, the more tinygrams can be sent

When there is congestion, so ACKs return more slowly, fewer tinygrams are sent



Nagle can reduce the number of segments significantly when the network is heavily loaded



Nagle can reduce the number of segments significantly when the network is heavily loaded

On the other hand, sometimes buffering up tinygrams is not a good idea: e.g., in a graphical interface over a network, each mouse movement becomes a tinygram. Buffering the segments would cause the cursor to jump erratically



Nagle can reduce the number of segments significantly when the network is heavily loaded

On the other hand, sometimes buffering up tinygrams is not a good idea: e.g., in a graphical interface over a network, each mouse movement becomes a tinygram. Buffering the segments would cause the cursor to jump erratically

Nagle can be turned off for such cases