Implementation of Transmission Control Protocol in Linux

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ABSTRACT
Transmission Control Protocol is the most used transmission layer protocol in the Internet. In addition to reliable and good performance in transmission between two nodes, it provides congestion control mechanism that is a major reason why Internet has not collapsed. Because of its complicated nature, implementations of it can be challenging to understand. This paper describes fundamental details of Transmission Control Protocol implementation in Linux kernel. Focus is on clarifying data structures and segments route through TCP stack.

1. INTRODUCTION
In May 1974 Vint Cerf and Bob Kahn published paper where they described an inter-networked protocol, which central control component was Transmission Control Program [3, 2]. Later it was divided into modular architecture and in 1981 Transmission Control Protocol (TCP), as it is know today, was specified in RFC 793 [8].

Today, TCP is the most used transmission layer protocol in the Internet [4] providing reliable transmission between two hosts through networks [8]. In order to gain good performance for communication, implementations of TCP must be highly optimized. Therefore, TCP is one of the most complicated components in Linux networking stack. In kernel 3.5.4, it consists of over 21000 lines of code under net/ipv4/-directory (all tcp*.c files together), while IPv4 consist of less than 13000 lines of code (all ip*.c files in the same directory). This paper explains the most fundamental data structures and operations used in Linux to implement TCP.

TCP provides reliable communication over unreliable network by using acknowledgment messages. In addition to provide resending of the data, TCP also controls its sending rate by using so-called 'windows' to inform the other end how much of data receiver is ready to accept.

As parts of the TCP code are dependent on network layer implementation, the scope of this paper is limited to IPv4 implementation as it is currently supported and used more widely than IPv6. However, most of the code is shared between IPv4 and IPv6, and tcp_ipv6.c is the only file related to TCP under net/ipv6/. In addition, TCP congestion control will be handled in a separate paper, so it will be handled very briefly. If other assumptions is made it is mentioned in the beginning of the related section.

Paper structure will be following: First section “Overview of implementation” will cover most important files and basic data structures used by TCP (tcp_sock, sk_buff), how data is stored inside these structures and how different queues are implemented, what timers TCP is using and how TCP sockets are kept in memory. Then a socket initialization and data flows through TCP are discussed. Section “Algorithms and optimizations” will handle logic of TCP state machine, briefly look into congestion control and explain what is TCP fast path.

2. OVERVIEW OF IMPLEMENTATION
In this section basic operation of TCP in Linux will be explained. It covers the most fundamental files and data structures used by TCP, as well as functions used when we are sending to or receiving from network.

The most important source files of implementation are listed in the Table 1. In addition to net/ipv4/, where most TCP files are located, there are also few headers located in include/net/ and include/linux/-directories. 1

<table>
<thead>
<tr>
<th>File</th>
<th>Description</th>
</tr>
</thead>
<tbody>
<tr>
<td>tcp.c</td>
<td>Layer between user and kernel space</td>
</tr>
<tr>
<td>tcp_output.c</td>
<td>TCP output engine. Handles outgoing data and passes it to network layer</td>
</tr>
<tr>
<td>tcp_input.c</td>
<td>TCP input engine. Handles incoming segments.</td>
</tr>
<tr>
<td>tcp_timer.c</td>
<td>TCP timer handling</td>
</tr>
<tr>
<td>tcp_ipv4.c</td>
<td>IPv4 related functions, receives segments from network layer</td>
</tr>
<tr>
<td>tcp_cong.c</td>
<td>Congestion control handler, includes also TCP Reno implementation</td>
</tr>
<tr>
<td>tcp_[reno</td>
<td>vegas</td>
</tr>
<tr>
<td>tcp.h</td>
<td>Main header files of TCP. struct tcp_sock is defined here. Note that there is tcp.h in both include/net/ and include/linux/</td>
</tr>
</tbody>
</table>

1Note that this paper is based on kernel version 3.5.3. In Linux 3.7, the new UAPI header file split moved some header files to new locations.
2.1 Data structures

Data structures are crucial sections of any software in order of performance and re-usability. As TCP is a highly optimized and remarkably complex entirety, robust understanding of data structures used is mandatory for mastering the implementation.

2.1.1 struct tcp_sock

struct tcp_sock (include/linux/tcp.h) is the core structure of TCP. It contains all the information and packet buffers for certain TCP connection. Figure 1 visualizes how this structure is implemented in Linux. Inside tcp_sock there is a few other, more general type of sockets. As a next, more general type of socket is always first member of socket type, can a pointer to socket be type-casted to other type of socket. This allows us to make general functions that handles with, for example, struct sock, even in reality pointer would be also a valid pointer to struct tcp_sock. Also depending on the type of the socket can different structure be as a first member of the certain socket. For example, as UDP is connection-less protocol, first member of struct udp_sock is struct inet_sock, but for struct tcp_sock first member must be struct inet_connection_sock, as it provides us features needed with connection-oriented protocols.

From Figure 1 it can be seen that TCP has many packet queues. There is receive queue, backlog queue and write queue (not in figure) under struct sock, and pre-queue and out-of-order queue under struct tcp_sock. These different queues and their functions are explained in detail in section 2.1.2.

struct inet_connection_sock (include/net/inet_connection_sock) is a socket type one level down from the tcp_sock. It contains information about protocol congestion state, protocol timers and the accept queue.

Next type of socket is struct inet_sock (include/net/inet_sock.h). It has information about connection ports and IP-addresses.

Finally there is general socket type struct sock. It contains two of TCP’s three receive queues, sk_receive_queue and sk_backlog, and also queue for sent data, used with retransmission.

2.1.2 Data queues

Incoming data queues are used as data storage before user reads the data to user space. These queues are implemented as double linked ring-list of struct sk_buffs (see section 2.1.3).

When user reads data from the socket, socket will be marked as being in use to avoid conflicts. However, incoming segments must be saved even when the socket is in use. Therefore, socket has several queues for incoming data: receive queue, pre-queue, and backlog queue. In addition to these, out-of-order queue is used as temporary storage for segments arriving out of order.

In the normal case when segment arrives and user is not waiting for the data, segment is processed immediately and the data is copied to the receive queue. Data will be copied to user’s buffer when application reads data from the socket.

If user is using blocking IO and the receive queue does not have as many bytes as requested, will the socket be marked as waiting for data. When the new segment arrives, it will be put to pre-queue and waiting process will be awakened. Then the data will be handled and copied to user’s buffer.

If user is handling segments (socket is marked as being in use) at the same time when we receive a new one, it will be put to the backlog queue, and user context will handle the segment after it has handled all earlier segments from other queues.

In addition to incoming data queues, there is also outgoing data queue known as write queue. It is implemented and used in the same way as incoming buffers. Segments will be put to write queue as user writes data to socket, and they will be removed when an acknowledgment arrives from the receiver.

Figure 4 visualizes use of receive, pre- and backlog-queues.

2.1.3 struct sk_buff

struct sk_buff (located in include/linux/skbuff.h) is used widely in the network implementation in Linux kernel. It is a socket buffer containing one slice of the data we are sending or receiving. In Figure 2 we see how data is stored inside structure. Data is hold in the continuous memory area surrounded by empty spaces, head and tail rooms. By having these empty spaces more data can be added to before or after current data without needing to copy or move it, and minimize risk of need to allocate more memory. However, if the data does not fit to space allocated, it will be fragmented to smaller segments and saved inside struct skb_shared_
Figure 1: Socket structures involved in TCP connection

All the data cannot be held in one large segment in the memory, and therefore we must have several socket buffers to be able to handle major amounts of data and to resend data segment that was lost during transmission to receiver. Because of that need of network data queues is obvious. In Linux these queues are implemented as double linked ring-lists of sk_buff structures (Figure 3). Each socket buffer has a pointer to the previous and next buffers. There is also special data structure to represent the whole list, known as struct sk_buff_head, that is used to indicate the first and the last members of ring list. More detailed information about the data queues is in section 2.1.2.

In addition data pointers, sk_buff also has pointer to owning socket, device from where data is arriving from or leaving by and several other members. All the members are documented in skbuff.h.

2.1.4 Hash tables
Hash table is a data structure that is used to map a given key to corresponding value.

Sockets are located in kernel’s hash table from where they are fetched when a new segment arrives or socket is otherwise needed. Main hash structure is struct inet_hashinfo (include/net/inet_hashtables.h), and TCP uses it as a type of global variable tcp_hashinfo located in net/ipv4/tcp_ipv4.c. struct inet_hashinfo has three main hash tables: One for sockets with full identity, one for bindings and one for listening sockets. In addition to that, full identity hash table is divided in to two parts: sockets in TIME_WAIT state and others.

As hash tables are more general and not only TCP specific part of kernel, this paper will not go into logic behind these more deeply.

2.1.5 Other data structures and features
There are also several other data structures that must be known in order to understand how TCP stack works. struct proto (include/net/sock.h) is a general structure presenting transmission layer to socket layer. It contains function pointers that are set to TCP specific functions in net/ipv4/tcp_ipv4.c, and applications function calls are eventually, through other layers, mapped to these.

struct tcp_info is used to pass information about socket state to user. Structure will be filled in function tcp_get_info(). It contains values for connection state (Listen, Established, etc), congestion control state (Open, Disorder, CWR, Recovery, Lost), receiver and sender MSS, rtt and various counters.

To provide reliable communication with good performance, TCP uses four timers: Retransmit timer, delayed ack timer, keep-alive timer and zero window probe timer. Retransmit, delayed ack and zero window probe timers are located in struct inet_connection_sock, and keep-alive timer can be found from struct sock (Figure 1). Although there is dedicated timer handling file net/ipv4/tcp_timer.c, timers are set and reset in several locations in the code as a result of events that occur.

2.2 Socket initialization
TCP functions available to socket layer are set to previously explained (section 2.1.5) struct proto in tcp_ipv4.c. This structure will be held in struct inet_protosw in af_inet.c, from where it will be fetched and set to sk->sk_proto when user does socket() call. During socket creation in the function inet_create() function sk->sk_prot->init() will be called, which points to tcp_v4_init_sock(). From there the real initialization function tcp_init_sock() will be called.

Address-family independent initialization of TCP socket occurs in tcp_init_sock() (net/ipv4/tcp.c). The function will
be called when socket is created with socket() system call. In that function fields of structure tcp_sock are initialized to default values. Also out of order queue will be initialized with skb_queue_head_init(), pre-queue with tcp_prequeue_init(), and TCP timers with tcp_init_xmit_timers(). At this point, state of the socket is set to TCP_CLOSE.

2.2.1 Connection socket

Next step to do when user wants to create a new TCP connection to other host is to call connect(). In the case of TCP, it maps to function inet_stream_connect(), from where sk->sk_prot->connect() is called. It maps to TCP function tcp_v4_connect().

tcp_v4_connect() validates end host address by using ip_route_connect() function. After that inet_hash_connect() will be called. inet_hash_connect() selects source port for our socket, if not set, and adds the socket to hash tables. If everything is fine, initial sequence number will be fetched from secure_tcp_sequence_number() and the socket is passed to tcp_connect().

tcp_connect() calls first tcp_connect_init(), that will initialize parameters used with TCP connection, such as maximum segment size (MSS) and TCP window size. After that tcp_connect() will reserve memory for socket buffer, add buffer to sockets write queue and passes buffer to function tcp_transmit_skb(), that builds TCP headers and passes data to network layer. Before returning tcp_connect() will start retransmission timer for the SYN packet. When SYN-ACK packet is received, state of socket is modified to ESTABLISHED, ACK is sent and communication between nodes may begin.

2.2.2 Listening socket

Creation of listening socket should be done in two phases. Firstly, bind() must be called to pick up port what will be listened to, and secondly, listen() must be called.

bind() maps to inet_bind(). Function validates port number and socket, and then tries to bind the wanted port. If everything goes fine function returns 0, otherwise error code indicating problem will be returned.

Function call listen() will become to function inet_listen(). inet_listen() performs a few sanity checks, and then calls function inet_csk_listen_start(), which allocates memory for socket accept queue, sets socket state to TCP_LISTEN and adds socket to TCP hash table to wait incoming connections.

2.3 Data flow through TCP in kernel

Knowing the rough route of incoming and outgoing segments through the layer is one of the most important part of TCP implementation to understand. In this section a roughly picture of it in most common cases will be given. Handling of all the cases is not appropriate and possible under the limits of this paper.

In this section it is assumed that DMA (CONFIG_NET_DMA) is not in use. It would be used to offload copying of data to dedicated hardware, thus saving CPU time. [1]

2.3.1 From the network

Figure 5 shows us a simplified summary about incoming data flow through TCP in Linux kernel.

In the case of IPv4, TCP receives incoming data from network layer in tcp_v4_rcv() (net/ipv4/tcp_ipv4.c). The function checks if packet is meant for us and finds the matching TCP socket from the hash table using IPs and ports as the keys. If the socket is not owned by user (user context is not handling the data), we first try to put the packet to pre-queue. Pre-queuing is possible only when user context is waiting for the data. If pre-queuing was not possible, we pass the data to tcp_v4_do_rcv(). There socket state is checked. If state is TCP_ESTABLISHED, data is passed to tcp_rcv_established(), and copied to receive queue. Otherwise buffer is passed to tcp_rcv_state_process(), where all the other states will be handled.

If the socket was not owned by user in function tcp_v4_do_rcv(), data will be copied to the backlog queue of the socket.
When user tries to read data from the socket (tcp_recvmsg()), queues must be processed in order. First receive queue, then data from pre-queue will be waited, and when the process ready to release socket, packets from backlog will be copied to the receive queue. Handling of the queues must be preserved in order to ensure that data will be copied to user buffer in the same order as it was sent.

Figure 4 visualizes the overall queuing process.

2.3.2 From the user

Figure 6 shows us a simplified summary about outgoing data flow through TCP in Linux kernel.

When user-level application writes data to TCP socket, first function that will be called is tcp_sendmsg(). It calculates size goal for segments and then creates sk_buff buffers of calculated size from the data, pushes buffers to write queue and notifies TCP output engine of new segments. Segments will go through TCP output engine and end up to tcp_transmit_skb().
tcp_write_xmit() takes care that segment is sent only when it is allowed to. If congestion control, sender window or Nagle’s algorithm [7] prevent sending, the data will not go forward. Also retransmission timers will be set from tcp_write_xmit(), and after data send, congestion window will be validated referring to RFC 2861 [5].
tcp_transmit_skb() builds up TCP headers and passes data to network layer by calling function queue_xmit() found from struct inet_connection_sock from member icsk_af_ops.

3. ALGORITHMS AND OPTIMIZATIONS

This section will go through a few crucial parts of implementation and clarify why these are important features to have and to work properly in a modern TCP implementation.

3.1 TCP state machine

There are several state machines implemented in Linux TCP. Probably most known TCP state machine is connection state machine, introduced in RFC 793 [8]. Figure 3.1 presents states and transitions implemented in kernel. In addition to connection state machine TCP has own state machine for congestion control.

Majority of TCP states are handled in tcp_rcv_state_process(), as it handles all the states except ESTABLISHED and TIME_WAIT. TIME_WAIT is handled in tcp_v4_rcv(), and state ESTABLISHED in tcp_rcv_established(). From the viewpoint of user, ESTABLISHED is the most important state as the actual data transmission happens in that state. Therefore, tcp_rcv_established() is the most interesting and also likely the most optimized function in the TCP implementation. Implementation of it is divided into two parts: slow and fast path (section 3.3).

As stated, TIME_WAIT is handled in tcp_v4_rcv(). Depending on return value of tcp_timewait_state_process, packet will be discarded, acked or processed again with a new socket (if the packet was SYN initializing a new connection). Implementation of function is very clean and easy to follow.

3.2 Congestion control

At first TCP did not have specific congestion control algorithms, and due to misbehaving TCP implementations Internet had first ‘congestion collapse’ in October 1988. Investigation on that led to first TCP congestion control algorithms described by Jacobson in 1988 [6]. However, it took almost 10 years before official RFC based on Jacobson’s research on congestion control algorithms came out [9].

Main file for TCP congestion control in Linux is tcp_cong.c. It contains congestion control algorithm database, functions to register and to active algorithm and implementation of TCP Reno. Congestion algorithm is linked to rest of the TCP stack by using struct tcp_congestion_ops, that has function pointers to currently used congestion control algorithm implementation. Pointer to the structure is found in struct inet_connection_sock (member icsk_ca_ops), see it at Figure 1.

Important fields for congestion control are located in struct tcp_sock (see section 2.1.1). Being the most important
variable, member `snd_cwnd` presents sending congestion window and `rcv_wnd` current receiver window. Congestion window is the estimated amount of data that can be in the network without data being lost. If too many bytes is sent to the network, TCP is not allowed to send more data before an acknowledgment from the other end is received.

As congestion control is out of scope of this paper, it will not be investigated more deeply.

### 3.3 TCP fast path

Normal, so-called slow path is a comprehensive processing route for segments. It handles special header flags and out-of-order segments, but because of that, it is also requiring heavy processing that is not needed in normal cases during data transmission.

Fast path is an TCP optimization used in `tcp_rcv_established()` to skip unnecessary packet handling in common cases when deep packet inspection is not needed. By default fast path is disabled, and before fast path can be enabled, four things must be verified: The out-of-order queue must be empty, receive window cannot be zero, memory must be available and urgent pointer has not been received. This four cases are checked in function `tcp_fast_path_check()`, and if all cases pass, will fast path be enabled in certain cases. Even after fast path is enabled, segment must be verified to be accepted to fast path.

TCP uses technique known as header prediction to verify segment to fast path. Header prediction allows TCP input machine to compare certain bits in the incoming segment’s header to check if the segment is valid for fast path. Header prediction ensures that there are no special conditions requiring additional processing. Because of this fast path is easily turned off by setting header prediction bits to zero, causing header prediction to fail always. In addition to pass header prediction, segment received must be next in order to be accepted to fast path.

### 4. CONCLUSION

Implementation of TCP in Linux is a complex and highly optimized to gain as high performance as possible. Because of that it is also time-consuming process to get into code level in kernel and understand TCP details. This paper described the most fundamental components of the TCP implementation in Linux 3.5.3 kernel.

### 5. REFERENCES