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S. Schuetz  
NEC  
N. Koutsianas  
L. Eggert  
Nokia  
W. Eddy  
Verizon  
Y. Swami  
Nokia  
K. Le  
NSN  
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**TCP Response to Lower-Layer Connectivity-Change Indications**  
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## Abstract

When the path characteristics between two hosts change abruptly, TCP can experience significant delays before resuming transmission in an efficient manner or TCP can behave unfairly to competing traffic. This document describes TCP extensions that improve transmission behavior in response to advisory, lower-layer connectivity-change indications. The proposed TCP extensions modify the local behavior of TCP and introduce a new TCP option to signal locally received connectivity-change indications to remote peers. Performance gains result from a more efficient transmission behavior and there is no difference in aggressiveness in comparison to a newly-started connection.

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## **1. Introduction**

The Transmission Control Protocol (TCP) [[RFC0793](#)] generally assumes that the end-to-end path between two hosts has characteristics that are relatively stable over the lifetime of a connection. Although TCP's congestion control algorithms [[RFC2581](#)] can adapt to changes to the path characteristics after several round-trip times, they fail to support efficient operation in the few round-trip times immediately after a significant path change. This is due to the granularity of TCP's sampling mechanisms. Significant changes to path connectivity include loss or reestablishment of connectivity, and drastic, abrupt changes in round-trip time (RTT) or available bandwidth. Connectivity changes that occur on such short time-scales are becoming more common, due to host mobility or intermittent network attachment.

This document describes a set of complementary TCP extensions that improve behavior when transmitting over paths whose characteristics can change on short time-scales. TCP implementations that support these extensions respond to receiving generic, link-technology-independent, per-connection connectivity-change indications from lower layers. A connectivity-change indication signals that the characteristics of the end-to-end path between the local node and its peer have changed in some undefined way. The response mechanisms proposed for TCP act on this information in a conservative fashion. The specific response depends on the current state of a connection when a connectivity-change indication is received.

It is important to note that this addition of response mechanisms to lower-layer information is following an established precedent. TCP and other transport protocols already react to information and signals from lower layers; the proposed connectivity-change indications thus extend an established interface between layers in the protocol stack. TCP measures the end-to-end path to implicitly derive network-layer information. TCP also directly reacts to network-layer signals delivered via ICMP, for example, "Port Unreachable" or the now-deprecated "Source Quench" [[RFC1122](#)]. Explicit Congestion Notification (ECN) [[RFC3168](#)] and Quick-Start [[RFC4782](#)] are other sources of network-layer information for which response mechanisms for TCP have been defined. Connectivity-change indications are yet another source of lower-layer information that TCP can use to improve its operation.

A second important point to note is that the TCP response mechanisms to connectivity-change indications are purely optional efficiency improvements. In the absence of connectivity-change indications, a TCP that implements these changes behaves identically to an unmodified TCP. When lower layers provide connectivity-change







Congestion control, RTT and path-MTU state gathered over an old path before the move generally have no meaning for the new path. Because TCP uses stale information when resuming transmission over the new path, it can be either too aggressive or highly inefficient. Similar conditions may be found when fail-overs occur for multihomed hosts through the shim6 protocol. Some background on the types of scenarios that the technology described in this document is designed to work within is found in [Appendix A](#).

TCP already forces a slow-start restart in some cases where the network state becomes unknown, such as after an idle period or heavy losses. A first part of the response specified in this document involves a similar return to initial slow-start state in response to connectivity-change indications that are received while a connection is transmitting in steady-state. Note that this behavior is more conservative than the standard TCP response or lack of response. Some performance gains with the proposed mechanisms are due to either avoiding overloading the new path, which typically incurs an RTT, or using slow-start to quickly detect new capacity far above the point where steady-state had previously been near.

A second response component improves TCP operation in the presence of temporary connectivity disruptions. These disruptions can occur independently of mobility events and, for example, may be due to insufficient wireless access coverage or nomadic computer use. Connectivity disruptions can severely decrease TCP performance. The main reason for this decrease is TCP's retransmission behavior after a connectivity disruption [[SCHUETZ](#)]. TCP uses periodic retransmission attempts in exponentially increasing intervals, which can unnecessarily delay retransmissions after connectivity returns. In the extreme case, TCP connections can even abort, if the disruption is longer than the TCP "user timeout". (Connection aborts are out of scope for this document but can be prevented by the TCP User Timeout Option [[I-D.ietf-tcpm-tcp-uto](#)].)

This second response action executes when receiving a connectivity-change indication while a connection is stalled in exponential back-off. It improves TCP retransmission behavior after connectivity is restored through an immediate speculative retransmission attempt [[footnote-1](#)]. Similar to the first response component, the second one also increases TCP performance through a more intelligent transmission behavior that uses periods of connectivity more efficiently. In comparison to startup of a new connection, it does not cause significant amounts of additional traffic and it does not change TCP's congestion control algorithms.

Finally, this draft specifies a third response component, which is a new TCP option that notifies the connection's remote peer of a





connectivity-change event detected locally. This is useful because connectivity-change indications typically require appropriate responses at both ends of a connection, but may only be received or detected by one end. The other parts of the response to a connectivity-change indication are independent of the indication's source (locally notified or remotely signaled) and depend only on the specific indication and the state of the connection for which it was received.

#### **4. Connectivity-Change Indications**

The focus of this document is on specifying TCP response mechanisms to lower-layer connectivity-change indications. This section briefly describes how different network- and shim-layer mechanisms underneath the transport layer may provide these connectivity-change indications to TCP. This section is included for clarification only; details on connectivity indication sources are out of scope of this document.

When lower layers detect a connectivity-change event, they generate corresponding connectivity-change indications. Lower-layer events that could trigger such an indication include (but are not limited to):

- o the IP address of the local outbound interface used for a given connection has changed, e.g., due to DHCP [[RFC2131](#)] or IPv6 router advertisements [[RFC2460](#)];
- o link-layer connectivity of the local outbound interface used for a given connection has changed, e.g., link-layer "link up" event [[RFC4957](#)];
- o the local outbound interface used for a given connection has changed, due to routing changes or link-layer connectivity changes at other interfaces (including tunnel establishment or teardown, e.g., in response to IKE events [[RFC4306](#)]);
- o a Mobile IP binding update has completed [[RFC3775](#)];
- o a HIP readdressing update has completed [[I-D.ietf-hip-mm](#)];
- o a path-change signal from the network has arrived (possible in theory, depends on network capabilities);
- o other notifications as defined by the IETF's Detecting Network Attachment (DNA) working group have occurred [[RFC4957](#)].

Note that the list above only describes some potential sources for



connectivity-change events. Other sources exist, but the details on when to generate such events are out of the scope of this document, which focuses on the TCP response mechanisms when such events are received.

## **5. TCP Response to Connectivity-Change Indications (CCIs)**

A TCP connection can receive a connectivity-change indication (CCI) either from its local stack ("local CCI") or through a new "connectivity-change indication TCP option" from its peer ("remote CCI"). [Section 5.1](#) specifies this new TCP option. In either case, upon reception of a CCI, the TCP RLCI (Response to Lower-layer Connectivity-change Indications) mechanisms defined in this document immediately re-probe path characteristics. They do this by either performing a speculative retransmission or by sending a single segment of new data or a pure ACK, depending on whether the connection is currently stalled in exponential back-off or transmitting in steady-state, respectively. A connection is "stalled in exponential back-off", if at least one segment was retransmitted due to a RTO expiration but has not been ACK'ed yet.

The remainder of this section first defines the format of the new CCI TCP option in [Section 5.1](#) and its processing in [Section 5.2](#). After that, the two TCP response mechanisms triggered by receiving CCIs - re-probing path characteristics and speculative retransmission - are described in [Section 5.3](#) and [Section 5.4](#).

The TCP RLCI mechanisms defined in this document depend on the TCP Timestamps option (TSopt) [[RFC1323](#)]. Consequently, it is REQUIRED that an end host that wishes to use the RLCI mechanisms for a TCP connection negotiate the use of TCP Timestamps options with its peer. If this negotiation fails, a host MUST NOT use the RLCI mechanisms for a connection. TCP Timestamps options are needed by the RLCI mechanisms during the following operations:

- o To re-probe the path characteristics after a connectivity-change indication. A host uses the TS Echo Reply (TSecr) field of a TCP Timestamps option to distinguish whether incoming ACKs are for segments that have been transmitted before or after CCI.
- o To identify a new remote CCI. A host uses the TS Value (TSval) field of an incoming TCP Timestamps option to distinguish a new remote CCI from the delayed reception of an old one. As a result, last remote CCI is defined as the one received with the highest TS Value.

[Section 5.2](#) and [Section 5.3](#) give more details about how the RLCI



mechanisms use TCP Timestamps options.

An implementation of the RLCI mechanisms defined in this document maintains nine new state variables per TCP connection. [[footnote-2](#)]

#### LOCAL\_CCI

It is a 1-bit counter, having an initial value of 0. It is used for distinguishing the existence of a new local CCI. It changes its value every time a new local CCI received from the local stack starts being processed.

#### REMOTE\_CCI

It holds a copy of the last CCI value advertised by the peer through a CCI TCP option. This is a 1-bit counter initialized to 0 and gets updated in response to remote CCIs according to the rules defined in [Section 5.2](#).

#### LOCAL\_CCI\_STATUS

It holds the status of the processing of local CCIs. It can have three possible values: LOCAL\_CCI\_IDLE (0), LOCAL\_CCI\_NEW (1), LOCAL\_CCI\_ECHO\_ACK (2). The initial value is LOCAL\_CCI\_IDLE.

#### REMOTE\_CCI\_STATUS

It holds the status of the processing of the last remote CCI advertised by the peer through a CCI TCP option. It can have two possible values: REMOTE\_CCI\_IDLE (0), REMOTE\_CCI\_ECHO (1). The initial value is REMOTE\_CCI\_IDLE.

#### LAST\_CCI\_TIME

It holds the local time when the last CCI (either local or remote) was received. It is updated every time either LOCAL\_CCI or REMOTE\_CCI is modified.

#### REMOTE\_CCI\_PEER\_TIME

This variable is used in order to distinguish new remote CCIs from the retransmissions of the past ones. It holds the TS Value (TSval) of the Timestamps option of the segment advertising the last remote CCI. It is initialized when receiving the first segment from the peer and it is updated every time REMOTE\_CCI is modified.

#### LOCAL\_CCI\_PEER\_ECHO\_TIME

This variable is used in order to distinguish the echo of a new local CCI from delayed retransmissions of echoes of older local CCIs. It holds the TS Value (TSval) of the Timestamps option of the segment that echoed the last local CCI. It is initialized when receiving the first segment from the peer and it is updated every time LOCAL\_CCI\_STATUS changes from LOCAL\_CCI\_NEW to



LOCAL\_CCI\_ECHO\_ACK.

CCI\_SNDMAX

Retains the highest sequence number transmitted when the most recent CCI (either local or remote) was received.

CCI\_CONTROLLED\_CWND

It is a Boolean variable that sets an additional condition controlling the increment of TCPs congestion window (CWND). Having an initial value of false, it is updated according to the rules defined in [Section 5.2](#).

### 5.1. Connectivity-Change Indication (CCI) TCP Option

Connectivity-change indications (CCIs) are generally asymmetric, i.e., they may occur or be detected by one end but not the other. The basic idea behind the CCI option is to signal the occurrence of local CCIs to the other end, in order to allow also the other end to respond appropriately. Note that this assumes that paths will generally be symmetric, meaning that a CCI received by one end for its path to the other end will imply that the characteristics of the reverse path have changed, too.

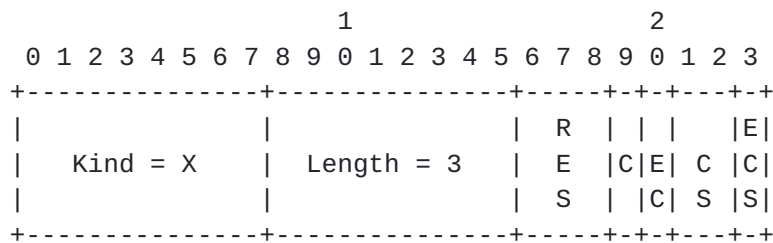


Figure 1: Format of the connectivity-change indication TCP option.

Figure 1 shows the format of the CCI option. It contains these fields:

Kind (8 bits)

The TCP option number X [[RFC0793](#)] allocated by IANA upon publication of this document (see [Section 8](#)).

Length (8 bits)

Length of the TCP option in octets [[RFC0793](#)]; its value MUST be 3.

RES (3 bits)

Reserved bits. The sender SHOULD set these to zero and the receiver MUST ignore them.





C (1 bit)

Current value of LOCAL\_CCI of the end sending the option.

EC (1 bit)

Echoed value of C, i.e., the current value of REMOTE\_CCI of the end sending the option.

CS (2 bit)

Current value of LOCAL\_CCI\_STATUS of the end sending the option.

ECS (1 bit)

Current value of REMOTE\_CCI\_STATUS of the end sending the option.

The CCI option contains two single-bit fields (C and EC) used to distinguish new CCIs from delayed retransmissions of past ones. It also contains some flags representing the status of each CCI processing. These flags are used for a 3-way handshake ensuring that both parties have been informed of a new CCI. At the beginning of a connection, LOCAL\_CCI and REMOTE\_CCI MUST be set to 0. LOCAL\_CCI\_STATUS and REMOTE\_CCI\_STATUS MUST be set to LOCAL\_CCI\_IDLE and REMOTE\_CCI\_IDLE, respectively.

A host actively opening a connection and wishing to use the CCI option for that connection MUST include a CCI option in its SYN segment with C := 0, CS := LOCAL\_CCI\_IDLE, EC := 0 and ECS := REMOTE\_CCI\_IDLE in order to advertise support for the TCP CCI option. A host receiving a SYN segment MUST NOT include a CCI option in its SYN-ACK or any subsequent segment, unless it has received a CCI option in the corresponding SYN. In case a host has received a CCI option in the SYN segment, it MUST echo that CCI option in its SYN-ACK segment, i.e., it MUST set C := 0, CS := LOCAL\_CCI\_IDLE, EC := 0 and ECS := REMOTE\_CCI\_IDLE. A host MUST NOT process any following CCI options unless one was included in both the SYN and SYN-ACK and both peers have enabled TCP Timestamps for the connection. [Section 5.2.1](#) and [Section 5.2.2](#) describe the processing rules in detail.

A host MUST send a CCI option in all outgoing segments whenever LOCAL\_CCI\_STATUS is not LOCAL\_CCI\_IDLE or REMOTE\_CCI\_STATUS is not REMOTE\_CCI\_IDLE (or both). A host MUST NOT send a CCI option when LOCAL\_CCI\_STATUS is LOCAL\_CCI\_IDLE and REMOTE\_CCI\_STATUS is REMOTE\_CCI\_IDLE, i.e., when the host is not currently processing any CCI. The only exceptions to that rule are SYN and SYN-ACK segments. Whenever sending any CCI option, C MUST be set to the current LOCAL\_CCI, EC MUST be set to the current REMOTE\_CCI, CS MUST be set to LOCAL\_CCI\_STATUS and ECS MUST be set to REMOTE\_CCI\_STATUS, respectively.



## **5.2. Generation and Processing of Connectivity-Change Indication TCP Options**

Processing of a connectivity-change indication can be separated into two parts:

1. Processing in "initiator" mode, i.e., when a host receives a local CCI and (reliably) forwards it to the other end through a CCI option.
2. Processing in "responder" mode, i.e., when a host that receives a remote CCI in a CCI option from the other end.

[Section 5.2.1](#) and [Section 5.2.2](#) describe the state machines at an initiator and a responder, respectively. Note that a single host can be both - initiator and responder - at the same time. This can happen if a local CCI occurs while processing for a remote CCI is ongoing, or vice versa.

The following events, conditions and actions are used in the definition of the two state machines:

Events:

E\_LOCAL\_CCI

Local end received a local CCI.

E\_REMOTE\_CCI

Local end received information about a remote CCI, i.e., received a TCP segment that includes a CCI option.

E\_SEGMENT\_SENT

Local end sent a TCP segment that includes the CCI option.

Conditions:

C\_NEW\_REMOTE\_CCI

A received CCI option signals a new remote CCI, i.e.,  $C \neq \text{REMOTE\_CCI}$ ,  $CS == \text{LOCAL\_CCI\_NEW}$  and the TSval of the Timestamps option of the received segment is greater than the current  $\text{REMOTE\_CCI\_PEER\_TIME}$  ( $\text{TSval} > \text{REMOTE\_CCI\_PEER\_TIME}$ ).

C\_ECHOED\_LOCAL\_CCI

A received CCI option echoes the last local CCI, i.e.,  $EC == \text{LOCAL\_CCI}$ ,  $ECS == \text{REMOTE\_CCI\_ECHO}$  and the TSval of the Timestamps option of the received segment is greater than the current  $\text{LOCAL\_CCI\_PEER\_ECHO\_TIME}$  ( $\text{TSval} > \text{LOCAL\_CCI\_PEER\_ECHO\_TIME}$ ).



#### C\_ECHOED\_REMOTE\_CCI

A received CCI option acknowledges that the peer has received the echo of its last local CCI, i.e.,  $C == \text{REMOTE\_CCI}$ ,  $CS == \text{LOCAL\_CCI\_ECHO\_ACK}$  and the TSval of the Timestamps option of the received segment is greater than the current  $\text{REMOTE\_CCI\_PEER\_TIME}$  ( $\text{TSval} > \text{REMOTE\_CCI\_PEER\_TIME}$ ).

#### Actions:

##### A\_TGL\_LOCAL\_CCI

Toggle LOCAL\_CCI.

##### A\_TGL\_REMOTE\_CCI

Toggle REMOTE\_CCI.

##### A\_REPROBE\_PATH

TCP discards all congestion control information gathered on the current path, initializes them to the defaults and re-probes path characteristics based only on the segments transmitted after this event, as described in [Section 5.3](#). In other words,  $\text{CCI\_CONTROLLED\_CWND} := 1$ ,  $\text{LAST\_CCI\_TIME} := \text{current local time}$ ,  $\text{CCI\_SNDMAX} := \text{highest sequence number transmitted so far}$  and the congestion control state (CWND and SS\_THRESH), round-trip time measurement (RTTM) state and RTO timer are reset to the initial values for a new connection. Additionally, if the connection is stalled in exponential back-off, TCP MUST act as if RTO had expired and start the speculative retransmission procedure described in [Section 5.4](#).

##### A\_FORCE\_SEND

Force transmission of a segment that MUST include a CCI option, in order to inform the other peer about the local CCI. If the connection is stalled in exponential back-off, this is taken care of by the speculative retransmission procedure described in [Section 5.4](#). If the connection is in steady-state and there is new data to be sent, TCP MUST immediately send a single segment of new data including a CCI option. If there is no new data to be sent, TCP MUST immediately send a pure ACK including a CCI option.

##### A\_UPD\_CCI\_PEER\_TIME

Set  $\text{REMOTE\_CCI\_PEER\_TIME}$  to the TSval value of the TCP Timestamps option of the received segment.

##### A\_UPD\_CCI\_PEER\_E\_TIME

Set  $\text{LOCAL\_CCI\_PEER\_ECHO\_TIME}$  to the TSval value of the TCP Timestamps option of the received segment.



### **5.2.1. Initiator Mode Processing**

This section describes the initiator mode processing of a TCP host implementing RLCI. In initiator mode, a host signals the occurrence of a local CCI to its peer, until the peer echoes reception of that CCI. After receiving the echo, the host needs to acknowledge the echo reception, resulting in a 3-way handshake. Figure 2 shows the corresponding state machine.

At the beginning of a connection, i.e., before the first local CCI occurs, LOCAL\_CCI is 0 and LOCAL\_CCI\_STATUS is LOCAL\_CCI\_IDLE. This remains the case until TCP receives a local CCI (E\_LOCAL\_CCI).

When that happens, TCP toggles LOCAL\_CCI (A\_TGL\_LOCAL\_CCI), sets LOCAL\_CCI\_STATUS := LOCAL\_CCI\_NEW, starts re-probing the new path (A\_REPROBE\_PATH) and forces a segment to be sent to the peer (A\_FORCE\_SEND).

Note that all subsequently transmitted segments MUST contain a CCI option until LOCAL\_CCI\_STATUS becomes LOCAL\_CCI\_IDLE. After the host receives the echo of the local CCI (C\_ECHOED\_LOCAL\_CCI), it updates LOCAL\_CCI\_PEER\_ECHO\_TIME (A\_UPD\_CCI\_PEER\_E\_TIME) and sets LOCAL\_CCI\_STATUS := LOCAL\_CCI\_ECHO\_ACK. The initiator remains in this state until it can send a segment with the CCI option (E\_SEGMENT\_SENT) that acknowledges reception of the CCI echo. At that time, it sets LOCAL\_CCI\_STATUS := LOCAL\_CCI\_IDLE.

The transition from LOCAL\_CCI\_IDLE to LOCAL\_CCI\_ECHO\_ACK occurs if a segment acknowledging the reception of a CCI echo is lost, and the initiator retransmits the echo acknowledgment.

When a local CCI occurs (E\_LOCAL\_CCI) while LOCAL\_CCI\_STATUS != LOCAL\_CCI\_IDLE, the host MUST ignore it and MUST NOT alter LOCAL\_CCI, because it is already processing another local CCI.





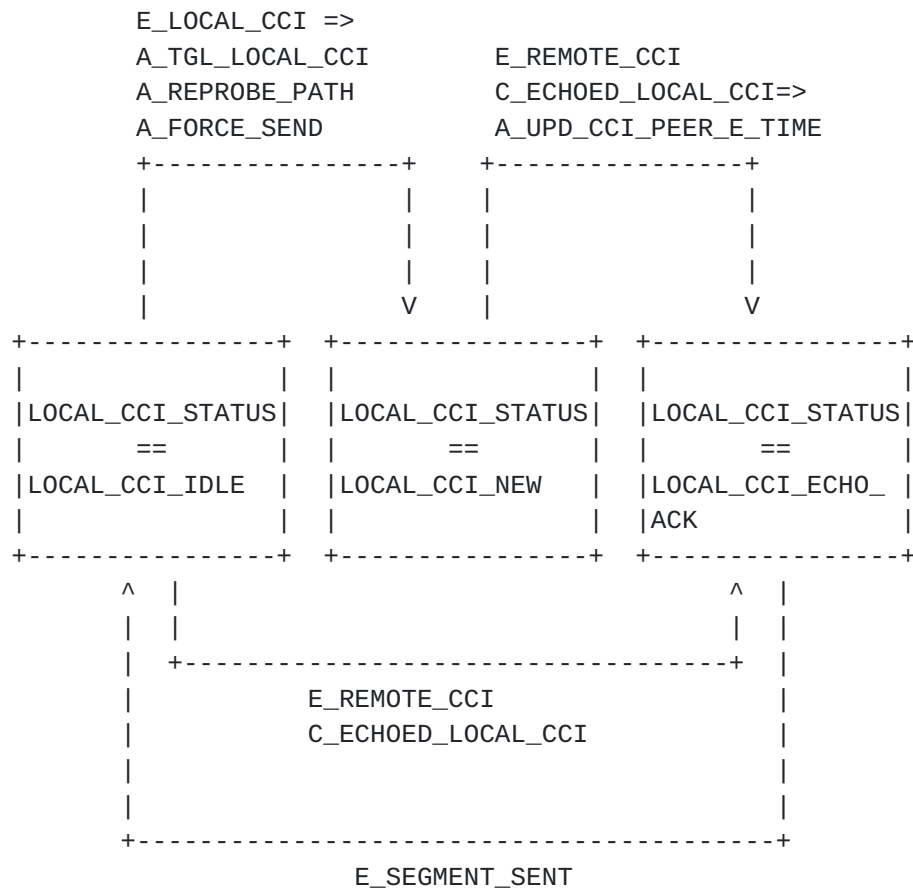


Figure 2: State machine for initiator processing.

### 5.2.2. Responder Mode Processing

This section describes the responder mode processing of CCIs for a TCP host implementing the CCI option. In responder mode, a host echoes the last received remote CCI to its peer, until it can be sure that the peer correctly received the echo. Figure 3 shows the corresponding state machine.

At the beginning of a connection, REMOTE\_CCI is 0 and REMOTE\_CCI\_STATUS is REMOTE\_CCI\_IDLE, i.e., the local host is not processing any remote CCIs.

When TCP receives a segment with a CCI option (E\_REMOTE\_CCI) signaling a new remote CCI (C\_NEW\_REMOTE\_CCI), it increments REMOTE\_CCI (A\_TGL\_REMOTE\_CCI), changes REMOTE\_CCI\_STATUS to REMOTE\_CCI\_ECHO, updates REMOTE\_CCI\_PEER\_TIME according to TSval (A\_UPD\_CCI\_PEER\_TIME), starts re-probing the new path (A\_REPROBE\_PATH) and forces a segment to be sent to the peer







If Path MTU Discovery (PMTUD) is in use, the PMTUD state MUST also be reset [[RFC1191](#)][RFC1981][[RFC4821](#)].

One difference to an initial slow-start is that after a CCI, the connection may have segments in flight towards the destination along a previous path. Therefore, after a CCI, TCP MUST ignore any ACKs received for data that was sent before the CCI and it MUST update the congestion window solely based on ACKs for data that was sent after the CCI occurred.

The mechanism used for distinguishing ACKs for data sent after a CCI occurred from ACKs for data sent before a CCI occurred uses TCP Timestamps options. When a host receives a new CCI (either local or remote), LAST\_CCI\_TIME MUST be set to the current local time, CCI\_SNDMAX MUST be set to the highest sequence number transmitted so far and CCI\_CONTROLLED\_CWND MUST be set to true.

While CCI\_CONTROLLED\_CWND == true, TCP MUST update the congestion window based only on inbound ACKs that contain a TS Echo Reply (TSecr) value greater than or equal to LAST\_CCI\_TIME. Any inbound ACK with a TS Echo Reply (TSecr) value less than LAST\_CCI\_TIME MUST NOT cause an update to the congestion window, even if it advances the window. If CCI\_CONTROLLED\_CWND is true and the host receives an ACK with a sequence number greater than or equal to CCI\_SNDMAX, CCI\_CONTROLLED\_CWND MUST be set to false and the congestion control algorithm MUST begin to process all ACKs normally, without checking their Timestamps options.

#### **[5.4.](#) Speculative Retransmission**

The basic idea behind the speculative retransmission is to allow TCP to resume stalled connections as soon as it receives an indication that connectivity to previously unreachable peers may have returned.

When a TCP connection receives a new CCI - either from the local stack or in a CCI TCP option from the peer - and is currently stalled in exponential back-off, it MUST immediately initiate the standard retransmission procedure, just as if the RTO for the connection had expired.

## **[6.](#) Discussion**

This section discusses some design choices of the RLCI mechanisms that can affect TCP performance under certain circumstances.



### **6.1. Triggered Segment Transmission during Steady-State**

A TCP stack that implements RLCI mechanisms and receives a local CCI immediately sends a TCP segment (A\_FORCE\_SEND) in order to inform the other end of the CCI and resets all path information (A\_REPROBE\_PATH). When TCP is stalled in exponential back-off, this is taken care of by the speculative retransmission procedure that is triggered by the CCI.

On the other hand, when TCP is in steady-state, it sends a new segment (A\_FORCE\_SEND) if there is any new data queued for transmission. As usual, the number of unacknowledged segments is limited by CWND. However, CWND has just been reset to its initial value. This means that there is a possibility that the transmission sends a segment that is outside the current congestion window. Although this behavior may appear to be aggressive, it is in fact as conservative as a newly starting connection, because only a single unacknowledged segment is sent along the path after CCI.

### **6.2. Impact of Packet Loss**

If a connection is in exponential back-off when a CCI occurs, TCP considers all unacknowledged segments to be lost and the speculative retransmission procedure immediately starts.

On the other hand, if the connection is in steady-state when a CCI occurs, TCP considers all unacknowledged segments to still be in flight and continues sending new data. Depending on what caused a CCI, four scenarios are possible that differ in what happens to segments and ACKs in flight:

1. All (or at least the vast majority of) segments and ACKs in flight reach their respective destinations, i.e., there are no losses. In this case, TCP acts as if a new connection had started and re-probes the new path.
2. Some of the ACKs in flight from the receiver to the sender are lost. In this case, TCP behaves exactly as above, because a cumulative ACK for the new segment sent along the path after the CCI acknowledges all the previous unacknowledged segments.
3. Some of the data segments in flight from the sender to the receiver are lost. In this case, the new data segment transmitted after the CCI causes a duplicate ACK. As this duplicate ACK does not cause TCP to send another data segment, the connection stalls and a RTO occurs. After RTO, the standard retransmission procedure takes place with SS\_THRESH equal to INITIAL\_WINDOW/2 (i.e., the minimum allowed). This disables slow





start and causes a severely decreased performance. A possible solution is to execute the speculative retransmission procedure after receiving a CCI even if the connection is in steady-state.

4. Some of the data segments and some of the ACKs that are in flight are lost. This case is similar to the previous one.

In all these cases, it is also possible that the round-trip time changes significantly after the CCI, reordering data segments and ACKs that are still in flight with ones sent after the CCI. These reorderings appear to TCP as losses, and may result in the connection experiencing one of the above cases even if there was no actual packet loss.

### **6.3. Use of Limited Transmit with RLCI**

As described in the previous section, when a connection is in steady-state, a connectivity-change indication (CCI) resets all path information of TCP and causes one new data segment to be sent. In case of significant data segment loss before a CCI, the new data segment transmitted after a CCI causes a duplicate ACK. As this duplicate ACK does not trigger TCP to send another data segment, the connection stalls and an RTO occurs.

Limited Transmit [[RFC3042](#)] can be used in case of packet loss in order to cause the transmission of three duplicate ACKs and trigger the fast retransmission procedure. As it must not cause an amount of outstanding data more than the congestion window plus two segments, it cannot always be used after a CCI due to the initialized CWND. If the connection has more outstanding data than INITIAL\_WINDOW plus two segments before a CCI, resetting of CWND to the initial value after CCI causes an amount of outstanding data greater than the new CWND plus two segments and disables Limited Transmit.

A modified Limited Transmit algorithm can be used in combination with RLCI:

If CCI\_CONTROLLED\_CWND is true:

The Limited Transmit Algorithm as described in [[RFC3042](#)] should be followed, but without checking the amount of outstanding data, i.e., if a TCP sender has previously unsent data queued for transmission it should transmit new data upon the arrival of the first two consecutive duplicate ACKs when the receiver's advertised window allows this transmission.



If CCI\_CONTROLLED\_CWND is false:

The Limited Transmit Algorithm as described in [\[RFC3042\]](#) should be followed unmodified.

When the fast retransmission procedure is triggered by the modified Limited Transmit after a CCI, SS\_THRESH is set to INITIAL\_WINDOW/2 (i.e., the minimum allowed) as CWND before fast retransmission was equal to INITIAL\_WINDOW. As a result, slow-start is disabled causing decreased TCP performance.

A minor modification can keep SS\_THRESH unmodified in the previous case, i.e., if CCI\_CONTROLLED\_CWND == true and CWND == INITIAL\_WINDOW, keep SS\_THRESH unmodified (having its initial value) upon the reception of the third duplicate ACK that triggers the fast retransmission procedure.

#### **6.4. Simultaneous Processing of Connectivity-Change Indications**

As mentioned in [Section 5.2.1](#), if a local CCI occurs (E\_LOCAL\_CCI) while LOCAL\_CCI\_STATUS != LOCAL\_CCI\_IDLE, the host MUST ignore it, because it is already processing another local CCI. As a result, only one local CCI at each end can be processed at the same time. Consequently, as every remote CCI at one end is triggered by a local CCI at the other end, only one remote CCI at each end can be processed at the same time.

On the other hand, if both hosts receive connectivity-change indications from their local stacks (local CCIs) at almost the same time, there is a possibility of simultaneous processing of local and remote CCIs at both ends. In that case, path re-probing is triggered twice at each end in a very short time that can be lower than RTT. As this does not improve TCP performance, it can be avoided by triggering the A\_REPROBE\_PATH action only if CCI\_CONTROLLED\_CWND == false.

#### **7. Security Considerations**

The only foreseen security considerations with the techniques presented in this document result from either an attacker's ability to spoof valid TCP segments with CCI options that seemingly indicate connectivity changes, or an attacker's ability to generate bogus CCIs locally. An attacker might produce a stream of such false indicators that could keep a connection in slow-start at the initial window. One possible defense against this type of attack is to rate-limit the response to CCIs (whether local or remote). This is also probably less serious than other attacks such an empowered adversary could perform, like resetting the connection or injecting data. A similar



effect could be achieved without the new CCI option by forging duplicate ACKs that would keep a sender in loss recovery. If both sets of IP addresses, port numbers, and sequence numbers are guessable for a connection, then the connection should employ other measures [[RFC4953](#)] for protection against spoofed segments.

## **8. IANA Considerations**

This section is to be interpreted according to [[I-D.narten-iana-considerations-rfc2434bis](#)].

This document does not define any new namespaces. It requests that IANA allocate a new 8-bit TCP option number for the CCI option from the registry maintained at <http://www.iana.org/assignments/tcp-parameters>.

## **9. Acknowledgments**

This draft combines and obsoletes [[I-D.swami-tcp-lmdr](#)] and [[I-D.eggert-tcpm-tcp-retransmit-now](#)]. The authors would like to thank Mark Allman, Marcus Brunner, Alfred Hoenes, Shashikant Maheshwari, Kacheong Poon, Juergen Quittek, Stefan Schmid and Joe Touch for their comments and suggestions on this draft as well as the two original drafts.

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#### Editorial Comments

[footnote-1] The authors have heard the idea of triggering retransmits based on connectivity events of directly-connected links being attributed to Phil Karn ("kick" operation in the KAQ9 TCP stack). A thread from the PILC mailing list in 2000 discusses some thoughts on this (<http://www.isi.edu/pilc/list/archive/0691.html>).

[footnote-2] Although this specification introduces eight new per-connection state variables, a preliminary implementation of an earlier revision of this mechanism [[I-D.swami-tcp-lmdr](#)] only required around a hundred lines of kernel code.

#### [Appendix A](#). Background: Classification of Connectivity Disruptions

Connectivity disruptions can occur in many different situations.



They can be due to wireless interference, movement out of a wireless coverage area, switching between access networks, or simply due to unplugging an Ethernet cable. Depending on the situation in which they occur, the implications of connectivity disruptions are different and must be handled appropriately. This section attempts to classify different types of connectivity disruptions and discusses their implications and impact on TCP.

Two main properties of connectivity disruptions affect how TCP reacts to them: their duration and whether the path characteristics have significantly changed after they end. This document distinguishes between "short" and "long" disruptions and "changed" and "unchanged" path characteristics. Note that these two categories are orthogonal to each other, i.e., four types of connectivity disruptions exist.

Connectivity disruptions are "short" for a given TCP connection, if connectivity returns before the RTO fires for the first time, i.e., when TCP is still in steady-state. In this case, standard TCP recovers lost data segments through Fast Retransmit and lost ACKs through successfully delivered later ACKs. [Appendix A.1](#) briefly describes this case.

Connectivity disruptions are "long" for a given TCP connection, if the RTO fires at least once before connectivity returns, i.e., when TCP is in exponential back-off. In this case, TCP can be inefficient in its retransmission scheme, as described in [Appendix A.2](#).

Whether or not path characteristics change when connectivity returns is a second important factor for TCP's retransmission scheme. Standard TCP implicitly assumes that path characteristics remain unchanged across short disruptions by performing Fast Retransmit using the path parameters collected before the disruption. For long disruptions, standard TCP is more conservative and performs slow-start, re-probing the path characteristics from scratch. However, the standard behavior can be inefficient due to when it is initiated.

These implicit assumptions can cause standard TCP to misbehave or perform inefficiently in some scenarios. Figure 4 illustrates the standard TCP behavior.



Short Duration < RTO	+-----+-----+	
	Fast Retransmit using currently collected path characteristics	Fast Retransmit using currently collected path characteristics
Long Duration ≥ RTO	+-----+-----+	
	Slow-start	Slow-start
+-----+-----+		
Unchanged Path Characteristics		Changed Path Characteristics

Figure 4: Standard TCP behavior.

#### [A.1.](#) Short Connectivity Disruptions

One common cause of short connectivity disruptions that result in a change of the end-to-end path characteristics is transparent network layer mobility, via protocols such as Mobile IP, NEMO, or HIP. These protocols generally hide mobility events from the transport layer, but cannot mask the resulting changes to the end-to-end path that established TCP connections transmit over.

Consider a Mobile IP scenario as shown in Figure 5. At time T, a mobile node MN attaches to access network Net-1, connected to the Internet through access router AR-1 and has the care-of address <Net-1, MN>. It establishes a TCP connection to the correspondent node CN. While MN attaches to AR-1, packets between CN and <Net-1, MN> follow PATH-1 (via Cloud-1 and AR-1). Assume that at some time T+1, MN moves and then attaches to Net-2, which is reachable through AR-2 with the care-of address <Net-2, MN>. While MN attaches to AR-2, all packets between CN and <Net-2, MN> follow PATH-2 (through Cloud-2 and AR-2).





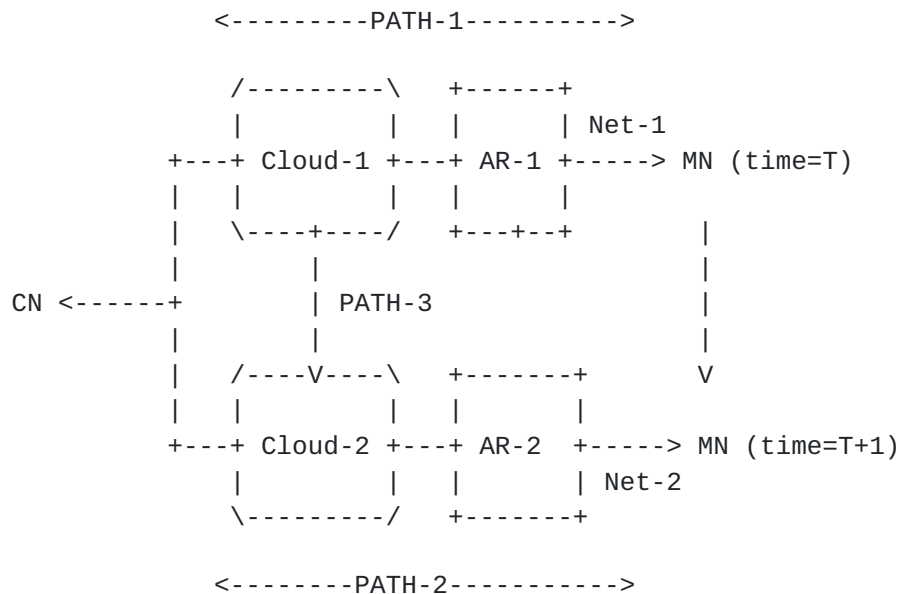


Figure 5: Mobility example.

During a transient disconnected period, MN may have disconnected from Net-1 and not yet attached to Net-2. Consequently, AR-1 may not be able to deliver packets to MN. This could result in a burst of packet losses. Several approaches for "fast" or "seamless" handovers exist that involve adding machinery to the ARs to buffer and redirect packets originally sent to Net-1 towards Net-2, rather than dropping them (e.g., [[KOODLI](#)]).

As long as MN remains in Net-1, standard congestion control algorithms [[RFC2581](#)] are sufficient. However, once MN moves from Net-1 to Net-2, two different scenarios are possible depending on network topology:

- o In the first scenario, with standard Mobile IPv4, all packets destined to <Net-1, MN> are dropped by AR-1 once MN has moved. Since the latency involved in establishing a new tunnel to the HA is on the order of the RTT ( $2 \times \text{RTT}$  in case of Mobile IPv6), roughly an entire window's worth of data and ACKs will be dropped by AR-1. Because of this burst loss, CN and MN are likely to incur expensive retransmission timeouts.
- o In the second scenario, with a fast handover mechanism in place, losses are masked through buffering and tunneling between routers AR-1 and AR-2. The exact sequence of buffering and forwarding between the ARs is not guaranteed to occur in a manner consistent with the available bandwidth of PATH-3 or conformant to TCP's clocking expectations. This can cause TCP's behavior over PATH-2 to be based on the unrelated properties of PATH-1 and PATH-3.



After attaching to Net-2, reception of stale ACKs (for data sent on PATH-1) will cause MN to incorrectly inflate its congestion window. These stale ACKs do not provide any indication of the congestion along PATH-2. CN's congestion window becomes similarly inflated by ACKs that MN sends for data segments redirected over PATH-3. If the congestion windows from PATH-1 are already too big for PATH-2, this can overload Net-2 or PATH-2, causing packet loss and timeouts.

On the other hand, if the available bandwidth along PATH-2 is greater than along PATH-1, and if the sender is in congestion avoidance, it will need potentially many RTTs before utilizing the available path capacity. This is due to relatively slow bandwidth increase during congestion avoidance caused by a stale SS\_THRESH. (See [\[EDDY\]](#) for details.)

## **[A.2.](#) Long Connectivity Disruptions**

For long disruptions, standard TCP performs slow-start after connectivity returns, because the retransmission timeout (RTO) has expired. This conservative strategy avoids overloading the new path. However, TCP's general exponential back-off retransmission strategy can time these slow-starts such that performance decreases.

When a long connectivity disruption occurs along the path between a host and its peer while the host is transmitting data, it stops receiving ACKs. After the RTO expires, the host attempts to retransmit the first unacknowledged segment. TCP implementations that follow the recommended RTO management proposed in [\[RFC2988\]](#) double the RTO after each retransmission attempt until it exceeds 60 seconds. This scheme causes a host to attempt to retransmit across established connections roughly once a minute. (More frequently during the first minute or two of the connectivity disruption, while the RTO is still being backed off.)

When the long connectivity disruption ends, standard TCP implementations still wait until the RTO expires before attempting retransmission. Figure 6 illustrates this behavior. Depending on when connectivity becomes available again, this can waste up to a minute of connectivity for TCPs that implement the recommended RTO management described in [\[RFC2988\]](#). For TCP implementations that do not implement [\[RFC2988\]](#), even longer connectivity periods may be wasted. For example, Linux uses 120 seconds as the maximum RTO by default.



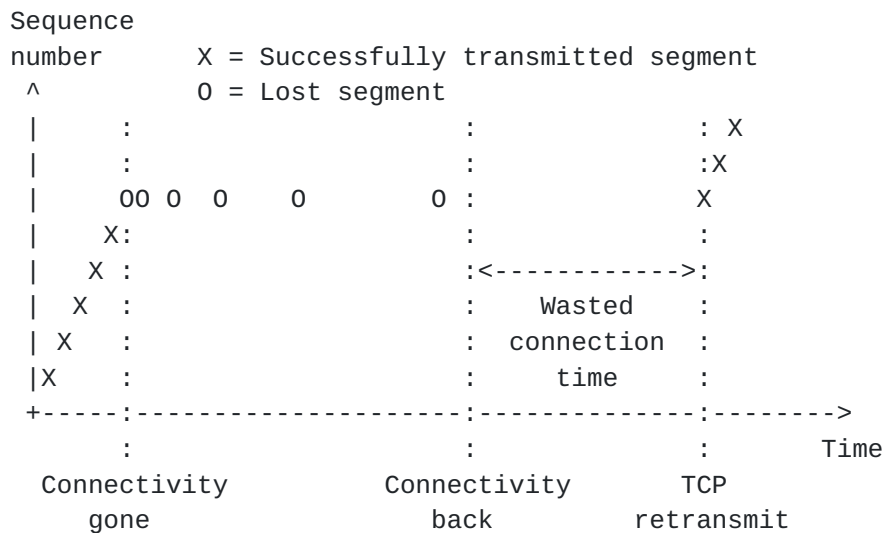


Figure 6: Standard TCP behavior in the presence of disrupted connectivity.

This retransmission behavior is not efficient, especially in scenarios where connectivity periods are short and connectivity disruptions are frequent [OTT]. Experiments show that TCP performance across a path with frequent disruptions is significantly worse, compared to a similar path without disruptions [SCHUETZ].

In the ideal case, TCP would attempt a retransmission as soon as connectivity to its peer was re-established. Figure 7 illustrates the ideal behavior.

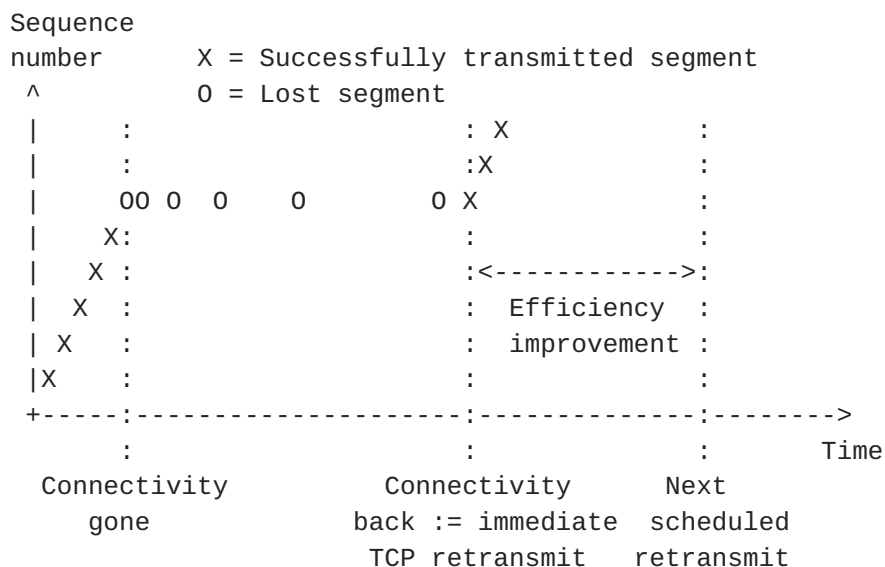


Figure 7: Ideal TCP behavior in the presence of disrupted connectivity









Authors' Addresses

Simon Schuetz  
NEC Laboratories Europe  
Kurfuerstenanlage 36  
Heidelberg 69115  
Germany

Phone: +49 6221 4342 165  
Email: [simon.schuetz@nw.neclab.eu](mailto:simon.schuetz@nw.neclab.eu)  
URI: <http://www.nw.neclab.eu>

Nikolaos Koutsianas  
Nokia Research Center

Email: [nkout@mobile.ntua.gr](mailto:nkout@mobile.ntua.gr)

Lars Eggert  
Nokia Research Center  
P.O. Box 407  
Nokia Group 00045  
Finland

Phone: +358 50 48 24461  
Email: [lars.eggert@nokia.com](mailto:lars.eggert@nokia.com)  
URI: [http://research.nokia.com/people/lars\\_eggert/](http://research.nokia.com/people/lars_eggert/)

Wesley M. Eddy  
Verizon Federal Network Systems  
NASA Glenn Research Center  
21000 Brookpark Road, MS 54-5  
Cleveland, OH 44135  
USA

Email: [weddy@grc.nasa.gov](mailto:weddy@grc.nasa.gov)



Yogesh Prem Swami  
Nokia Research Center, Dallas  
955 Page Mill Road  
Palo Alto, California 94304  
USA

Phone: +1 972 374 0669  
Email: yogesh.swami@nokia.com

Khiem Le  
Nokia Siemens Networks  
6000 Connection Drive  
Irving, TX 75039  
USA

Phone: +1 972 342 3502  
Email: khiem.le@nsn.com



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