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Examples call flow in race condition on Session Initiation Protocol
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Abstract

This document gives examples of the Session Initiation Protocol (SIP) call flows in race condition. Call flows in race condition are confusing, and this document shows the best practice to handle them. The elements in these call flows include SIP User Agents and SIP Proxies. Call flow diagrams and message details are shown.

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1. Overview

The call flows shown in this document were developed in the design of a SIP IP communications network. These examples are of race condition, which stems from the dialog state transition mainly established by INVITE.

When implementing SIP, various complex situations may arise. Therefore, it will be helpful to provide implementors of the protocol with examples of recommended terminal and server behavior.

This document clarifies SIP UA behaviors when messages cross each

other as race conditions. By clarifying operation under race conditions, inconsistent interpretations between implementations are

avoided and interoperability is expected to be promoted.

It is the hope of the authors that this document will be useful for SIP implementors, designers, and protocol researchers and will help them achieve the goal of a standard implementation of RFC 3261 [1].

These call flows are based on the version 2.0 of SIP defined in RFC 3261 [1] with SDP usage described in RFC 3264 [2].

The key words "MUST", "MUST NOT", "REQUIRED", "SHALL", "SHALL NOT", "SHOULD", "SHOULD NOT", "RECOMMENDED", "MAY", and "OPTIONAL" in this document are to be interpreted as described in BCP 14, RFC 2119 [4].

1.1 General Assumptions

A number of architecture, network, and protocol assumptions underlie the call flows in this document. Note that these assumptions are not requirements. They are outlined in this section so that they may be taken into consideration and help understanding the call flow examples.

These flows do not assume specific underlying transport protocols such as TCP, TLS, and UDP. See the discussion in RFC 3261 [1] for details on the transport issues for SIP.

1.2 Legend for Message Flows

Dashed lines (---) and slash lines (/,\) represent signaling messages that are mandatory to the call scenario. (X) represents crossover of signaling messages. (->x,x<-) indicate that the packet is lost. The arrow indicate the direction of message flow. Double dashed lines (===) represent media paths between network elements.

Messages are identified in the Figures as F1, F2, etc. These numbers are used for references to the message details that follow the Figure.
Comments in the message details are shown in the following form:

```
/* Comments.  */
```

1.3 SIP Protocol Assumptions

This document does not prescribe the flows precisely as they are shown, but rather illustrates the principles for best practice. They are best practice usages (orderings, syntax, selection of features for the purpose, or handling of error) of SIP methods,

headers and parameters. NOTE: The flows in this document must not

be copied as they are by implementors because additional characteristics were incorporated into the document for ease of explanation. To sum up, the procedures described in this document represent well-reviewed examples of SIP usage, which are best common practice according to IETF consensus.

For simplicity in reading and editing the document, there are a number of differences between some of the examples and actual SIP messages. For instance, Call-IDs are often repeated, CSeq often begins at 1, header fields are usually shown in the same order, usually only the minimum required header field set is shown, and other headers which would usually be included such as Accept, Allow, etc are not shown.

Actors:

Element	Display Name	URI	IP Address
User Agent	Alice	sip:alice@atlanta.example.com	192.0.2.101
User Agent	Bob	sip:bob@biloxi.example.com	192.0.2.201
User Agent	Carol	sip:carol@chicago.example.com	192.0.2.202
Proxy Server		ss.atlanta.example.com	192.0.2.111

2. The Dialog State Machine for INVITE dialog usage

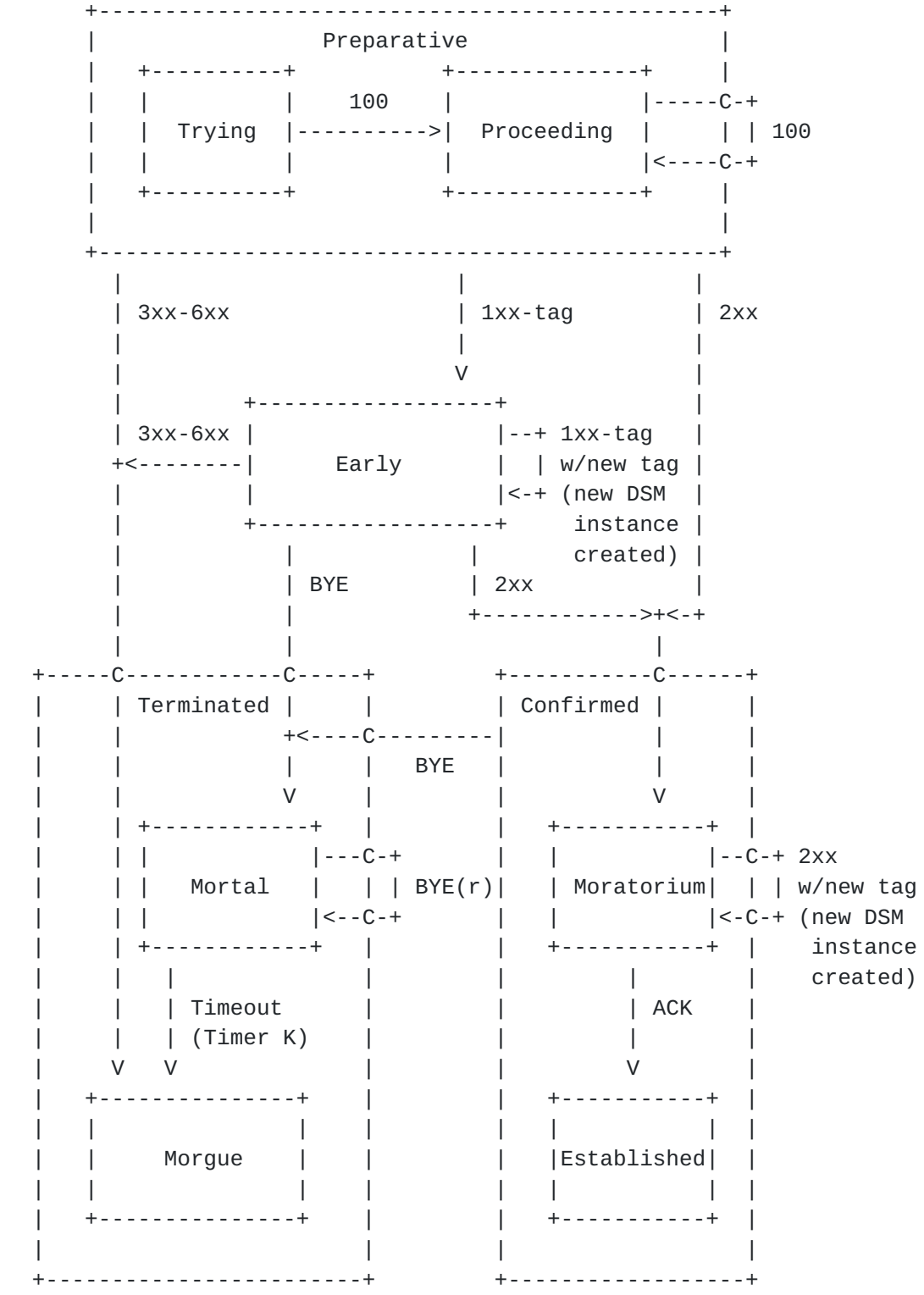
Race conditions are generated when the dialog state of the receiving side differs from that of the sending side.

For instance, a race condition occurs when UAC (User Agent Client) sends a CANCEL in the Early state while UAS (User Agent Server) is transiting from the Early state to the Confirmed state by sending a 200 OK to ini-INVITE.

The DSM (dialog state machine) for the INVITE dialog usage is presented as follows to help understanding UA's behavior in race conditions.

The DSM clarifies UA's behavior by subdividing some internal states showed in the FSM (Finite State Machine) for dialog state of the dialog-package [7], without changing the states of the dialog, "early", "confirmed", and "terminated" shown in RFC3261 [1]. The Preparative state is put before the Early state, which includes the Trying and Proceeding states. The Confirmed state is subdivided into two substates, the Moratorium and Established states and the Terminated state is subdivided into the Mortal and Morgue states.

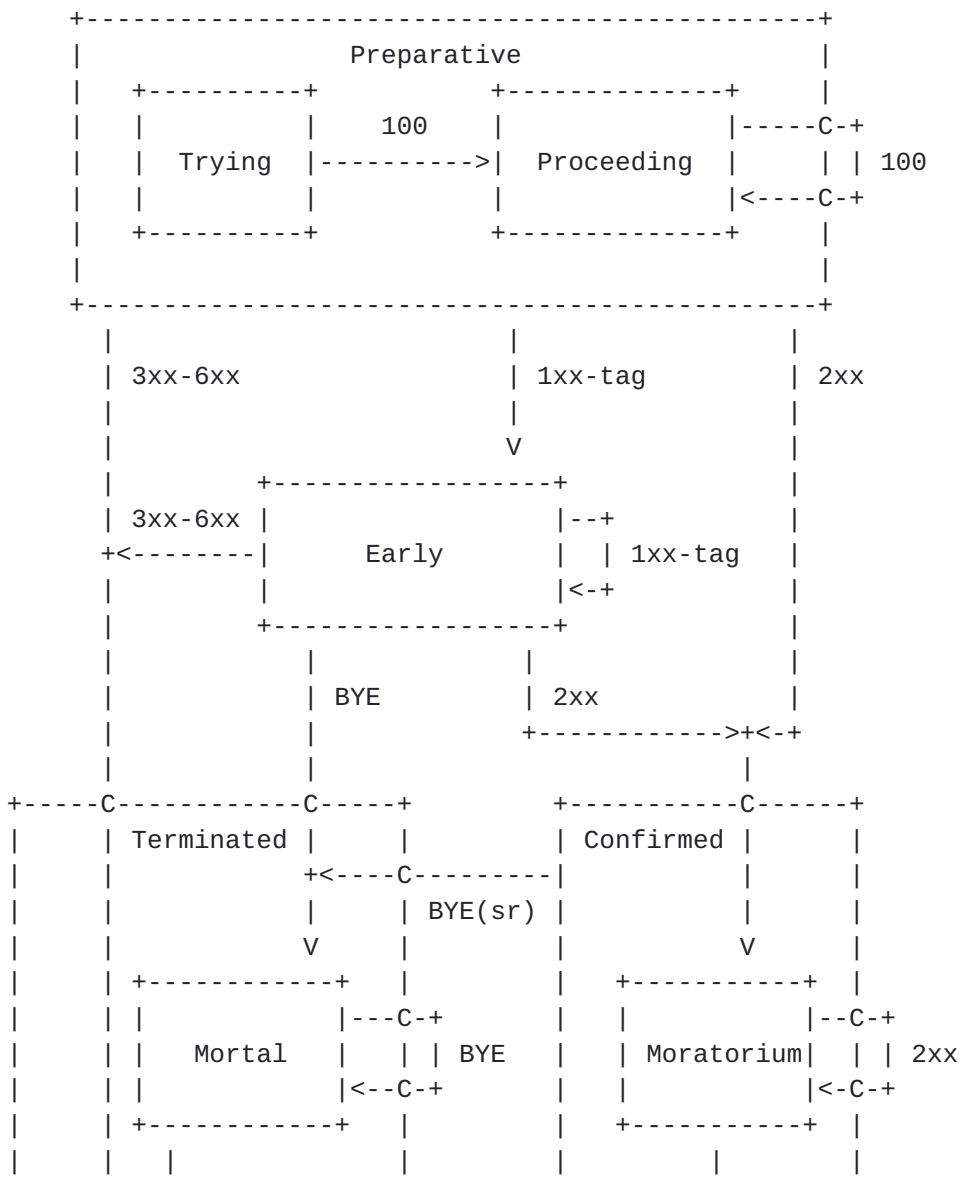
Below are the DSMs for UAC and UAS respectively.



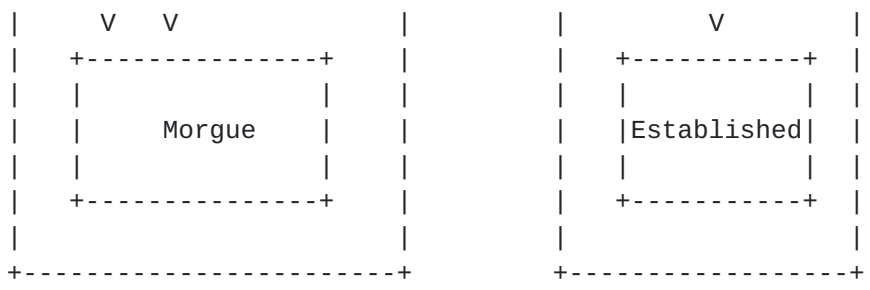
(r): indicates only reception is allowed.
Where (r) is not indicated, response means receive, request means send.

Figure 1. DSM for INVITE dialog usage (UAC)

Figure 1 represents the UAC's DSM for the INVITE dialog usage. UAC MAY send a BYE in the Early state, even though this behavior is NOT RECOMMENDED. The BYE sent in the Early state terminates the Early dialog with a specific To-tag. That is, when a proxy is performing forking, the BYE is only able to terminate the Early dialog between a particular UA. If UAC wants to terminate all Early dialogs instead of that with a particular UA, it needs to send CANCEL, not BYE. Moreover, until UAC receives a final response and terminates the INVITE transaction, the UAC MUST be prepared to establish a dialog by receiving a new response to the INVITE even though it had sent a BYE and terminated the dialog (see Appendix A).




| | | Timeout | | | ACK |
| | | (Timer J) | | | |



(sr): indicates that both sending and reception are allowed.
Where (sr) is not indicated, response means send,
request means receive.

Figure 2. DSM for INVITE dialog usage (UAS)

Figure 2 represents UAS's DSM for the INVITE dialog usage. The figure does not illustrate the state transition related to CANCEL request. CANCEL request does not cause a dialog state transition. However, the UAS terminates the dialog and triggers the dialog transition by sending 487 immediately after the reception of the CANCEL. Considering this, the behavior upon the reception of the CANCEL request is further explained in the Appendix C.

Following are UA's behaviors in each state.

Preparative (Pre): The Preparative state is a state until the Early dialog is established by sending and receiving a provisional response with To-tag after an ini-INVITE is sent and received. The dialog has not existed yet in the Preparative state. The dialog state transits from the Preparative to the Early by sending or receiving a provisional response with To-tag. Moreover, if UA sends or receives a 2xx response, the dialog state transit to the Moratorium state which is a substate of the Confirmed state.
In addition, if UA sends or receives a 3xx-6xx response the dialog state transit to the Morgue state which is a substate of the Terminated state. Sending an ACK for a 3xx-6xx response and retransmissions of 3xx-6xx are not expressed on this DSM because they are sent by the INVITE transaction.

Trying (Try): The Trying state is a substate of the Preparative state and inherits the behavior of the superstate. The Trying state is started by sending and receiving of an ini-INVITE. It transits to the Proceeding state by sending or receiving a 1xx (usually 100 trying) without To-tag. UAC may retransmit an INVITE on transaction layer and must not send a CANCEL request. UAS may send a 1xx-6xx response.

Proceeding (Pro): The Proceeding state is a substate of the

Preparative state and inherits the behavior of the superstate. Dialog becomes the Proceeding state if a dialog in the Trying state sends or receives a 1xx without To-tag (usually 100 trying). UAC may send CANCEL, and UAS may send a 1xx-6xx response in the Proceeding state.

Early (Ear): The early dialog is established by sending or receiving a provisional response with To-tag. The early dialog exists though the dialog does not existed in this state yet. The dialog state transits from the Early to Moratorium state, a substate of the Confirmed state, by sending or receiving a 2xx response. In addition, the dialog state transits to the Morgue state, a substate of the Terminated state, by sending and receiving a 3xx-6xx response. Sending an ACK for a 3xx-6xx response and retransmissions of 3xx-6xx are not expressed on this DSM because they are automatically processed on transaction layer and don't influence the dialog state. UAC may send CANCEL in the Early state. UAC may send BYE (although it is not recommended). UAS may send a 1xx-6xx response. Sending or receiving of a CANCEL request does not have direct influences on dialog state. The UA's behavior upon the reception of the CANCEL request is further explained in the Appendix C.

Confirmed (Con): Sending or receiving of a 2xx final response establishes a dialog. Dialog exists in this state. The BYE request the changes state from the Confirmed to Mortal state, a substate of the Terminated state. The Confirmed has two substates, the Moratorium and Established state, which are different in messages UAs are allowed to send.

Moratorium (Mora): The Moratorium state is a substate of the Confirmed state and inherits the behavior of the superstate. The Moratorium state transits to the Established state by sending or receiving an ACK request. UAC may send ACK and UAS may send a 2xx final response.

Established (Est): The Established state is a substate of the Confirmed state and inherits the behavior of superstate. Both caller and callee may send various messages which influences a dialog. Caller supports the transmission of ACK for a retransmission of a 2xx response to an ini-INVITE.

Terminated (Ter): The Terminated state is divided into two substates, the Mortal and Morgue states, to cover the behavior when a dialog is being terminated. In this state, UAs hold information about¡¡the dialog which is being terminated. The Confirmed state transits to the Mortal state, a substate of the

Terminated state, by sending or receiving a BYE request.

Mortal (Mort): Caller and callee becomes Mortal state by sending or receiving a BYE. UA MUST NOT send any new requests since there is no dialog. (Here the new requests do not include ACK for 2xx and BYE for 401 or 407 as further explained in the Appendix D below.)
In this state, only BYE or its response can be handled, and no other messages can be received. This is because the use case is taken into consideration that BYE is sent by both a caller and a callee to exchange reports about the session when it is being terminated. Therefore, UA possesses dialog information for internal process but dialog shouldn't exist outwardly. The UA stops managing its dialog state and changes it to the Morgue state, when the BYE transaction is finished by timer (Timer F or Timer K for UAC. Timer J for UAS).

Morgue (Morg): Dialog does not exist any more in this state. Sending or receiving of a signal which influences a dialog is not performed. (A dialog is literally terminated.)

3. Race condition

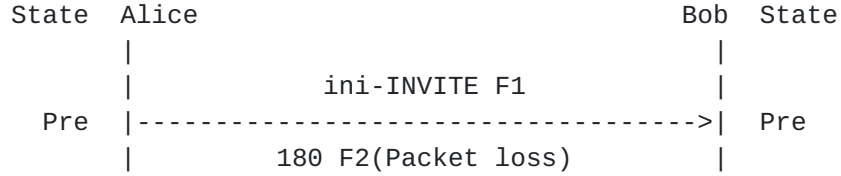
This section details race condition between two SIP UAs, Alice and Bob. Alice (sip:alice@atlanta.example.com) and Bob (sip:bob@biloxi.example.com) are assumed to be SIP phones or SIP-enabled devices.
Only significant signals are illustrated. Dialog state transitions caused by sending and receiving of SIP messages as well as '*race*', which indicates race condition are shown. (For abbreviations for the dialog state transitions, refer to Chapter 2.)
'*race*' indicates the moment when a race condition occurs.

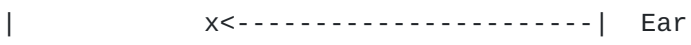
Examples of race conditions are shown below.

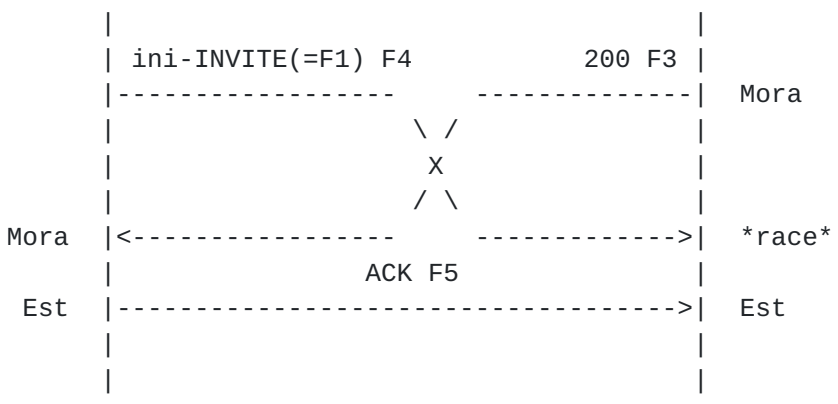
3.1 Receiving message in the Moratorium State

This section shows some examples of call flow in race condition when receiving the message from other states in the Moratorium state.

3.1.1 Receiving Initial INVITE retransmission (Trying state) in Moratorium state




x<----------| Ear



This scenario illustrates the race condition which occurs when UAS receives a Preparative message in the Moratorium state. All provisional responses to the initial INVITE (ini-INVITE F1) are lost, and UAC retransmits an ini-INVITE (F4). At the same time as retransmission, UAS generates a 200 OK (F3) to the ini-INVITE and it terminate an INVITE server transaction, according to Section 13.3.1.4 of RFC3261 [1].
However, it is reported that terminating an INVITE server transaction by 200 OK is a SIP bug. (http://bugs.sipit.net/, #769)
Therefore, the INVITE server transaction is not terminated at F3, and the F4 MUST be properly handled as a retransmission.
(UAs that do not deal with this bug still need to recognize the dialog relying on its From-tag and Call-ID, and the retransmitted request relying on the CSeq header field value even though it does not match the transaction.)
In RFC3261 [1], it is not specified whether UAS retransmits 200 to the retransmission of ini-INVITE. Considering the retransmission of 200 triggered by timer (TU keeps retransmitting 200 based on T1 and T2 until it receives an ACK), according to Section 13.3.1.4 of RFC3261 [1], it seems unnecessary to retransmit 200 when the UAS receives the retransmission of ini-INVITE. (For implementation, it does not matter if the UAS sends the retransmission of 200, since the 200 does not cause any problem.)

Message Details

F1 INVITE Alice -> Bob

F2 180 Ringing Bob -> Alice
/* 180 response is lost and does not reach Alice. */

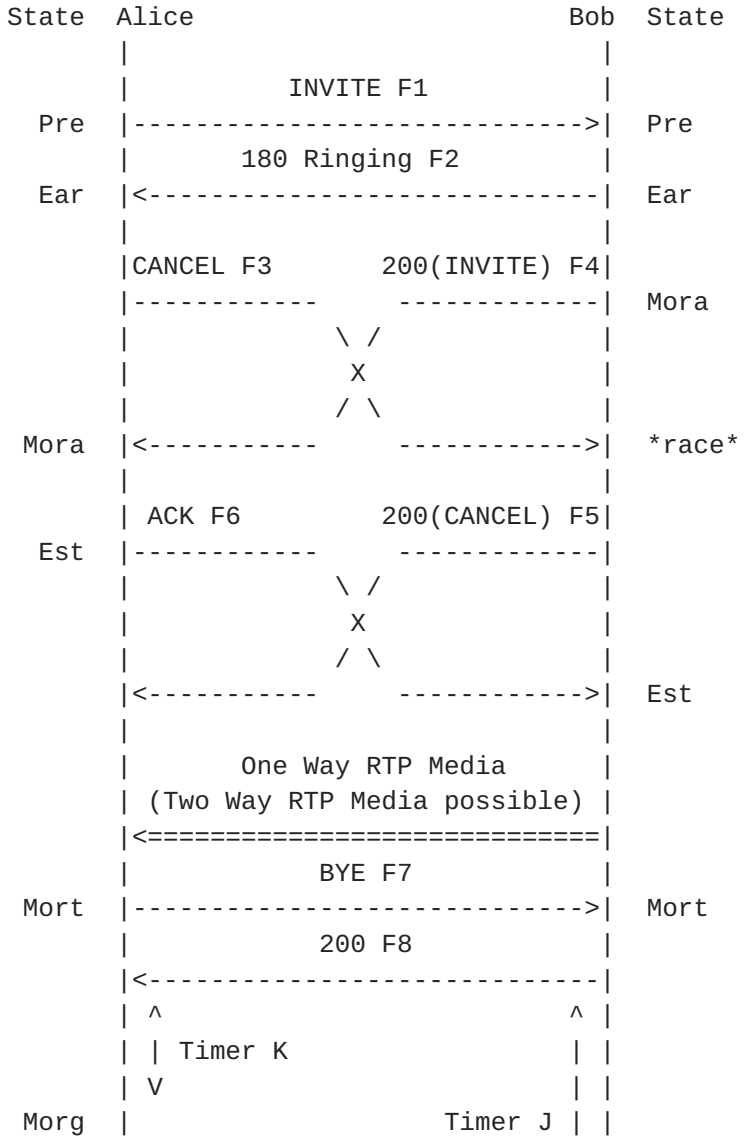
F3 200 OK Bob -> Alice
/* According to 13.3.1.4 of RFC3261, an INVITE server transaction is terminated at this point. However, this has been reported as a

SIP bug, and the UAS MUST correctly recognize the ini-INVITE (F4) as a retransmission. */

F4 INVITE (retransmission) Alice -> Bob
/* F4 is a retransmission of F1. They are exactly the same INVITE request. For UAs do not deal with the bug reported in #769 (an INVITE server transaction is terminated by 200 to INVITE), this request does not match the transaction as well as the dialog since it does not have a To-tag.
However, Bob have to recognize the retransmitted INVITE correctly, without treating it as a new INVITE. */

F5 ACK Alice -> Bob

3.1.2 Receiving CANCEL (Proceeding or Early state) in Moratorium state




V
Morg

This scenario illustrates the race condition which occurs when UAS receives an Early message, CANCEL, in the Moratorium state. Alice sends a CANCEL and Bob sends a 200 OK response to the initial INVITE message at the same time. As described in the previous section, according to RFC3261, an INVITE server transaction is supposed to be terminated by a 200 response, but this has been reported as a bug #769.
This section describes a case in which an INVITE server transaction is not terminated by a response to the CANCEL request. In this case, there is an INVITE transaction which the CANCEL request matches, so a 200 response is sent to the request. This 200 response simply means that the next hop received the CANCEL request (Successful CANCEL (200) does not mean an INVITE failure). When UAS does not deal with #769, UAC MAY receive a 481 response for CANCEL since there is no transaction which the CANCEL request matches. This 481 simply means that there is no matching INVITE server transaction and CANCEL is not sent to the next hop.
Regardless of the success/failure of the CANCEL, Alice checks the final response to INVITE, and if she receives 200 to the INVITE request she immediately sends a BYE and terminates a dialog. (Section 15, RFC3261 [1])
From the time F1 is received by Bob until the time that F8 is sent by Bob, media may be flowing one way from Bob to Alice. From the time than an answer is received by Alice from Bob there is the possibility that media may flow from Alice to Bob as well. However, once Alice has decided to cancel the call, she presumably will not send media, so practically speaking the media stream will remain one way.

Message Details

F1 INVITE Alice -> Bob

F2 180 Ringing Bob -> Alice

F3 CANCEL Alice -> Bob
/* Alice sends a CANCEL in the Early state. */

F4 200 OK (INVITE) Bob -> Alice
/* Alice receives a 200 to INVITE (F1) in the Moratorium state.
Alice has the potential to send as well as receive media,
but in practice will not send because there is an intent
to end the call. */

F5 200 OK (CANCEL) Bob -> Alice
/* 200 to CANCEL simply means that the CANCEL was received.

The 200 response is sent, since this document deals with the

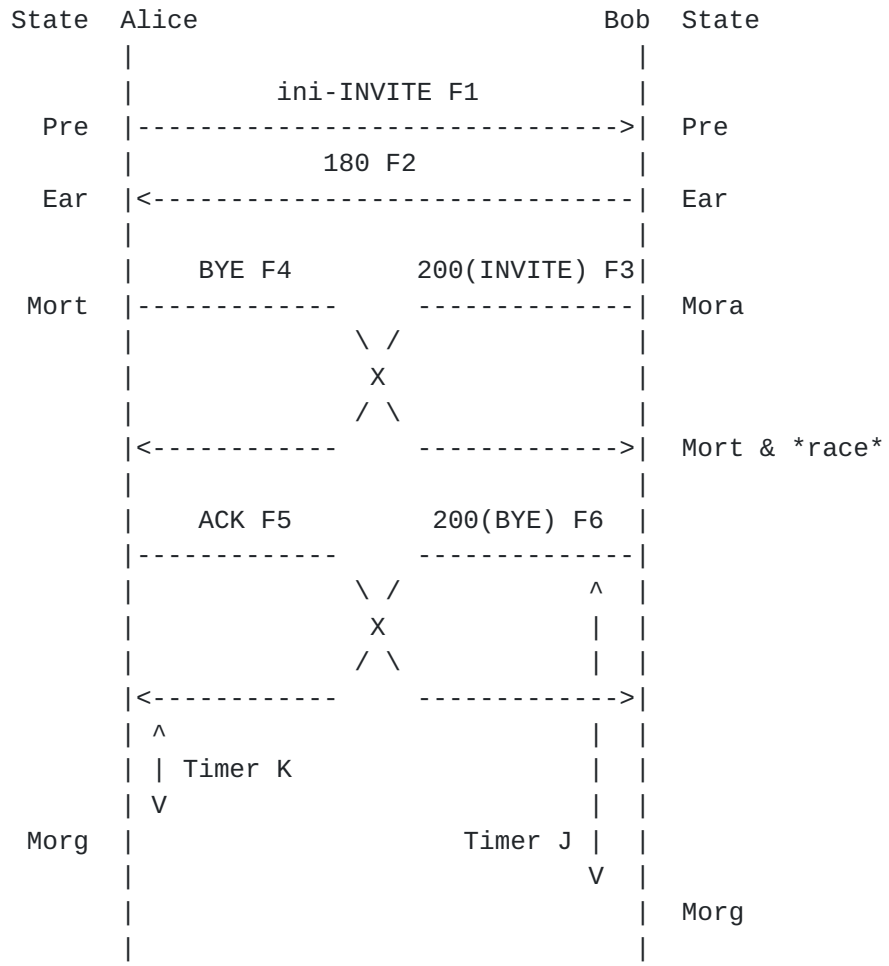
bug reported in #769. When an INVITE server transaction is terminated as the procedure stated in RFC3261, UAC MAY receive 481 response instead of 200. */

F6 ACK Alice -> Bob
/* INVITE is successful, and the CANCEL becomes invalid. Bob establishes RTP streams.
However, the next BYE request immediately terminates the dialog and session. */

F7 BYE Alice -> Bob

F8 200 OK Bob -> Alice

3.1.3 Receiving BYE (Early state) in Moratorium state



This scenario illustrates the race condition which occurs when UAS

receives an Early message, BYE, in the Moratorium state. Alice sends

a BYE in the Early state and Bob sends a 200 OK response to the initial INVITE request at the same time. Bob receives the BYE in the Confirmed dialog state though Alice sent the request in the Early state (As explained in Section 2, this behavior is NOT RECOMMENDED). The BYE functions normally even if it is received after the INVITE transaction termination because BYE differs from CANCEL, and is sent not to the request but to the dialog. Alice gets into the Mortal state on receiving the BYE response, and remains Mortal until the Timer K timeout occurs. In the Mortal state, UAC does not establish a session, even though it receives a 200 response to INVITE. Even so, the UAC sends an ACK to 200 for the completion of INVITE transaction. The ACK is always sent to complete the three-way handshake of INVITE transaction (Further explained in the Appendix D below).

Message Details

F1 INVITE Alice -> Bob

F2 180 Ringing Bob -> Alice

F3 200 OK (ini-INVITE) Bob -> Alice

F4 BYE Alice -> Bob

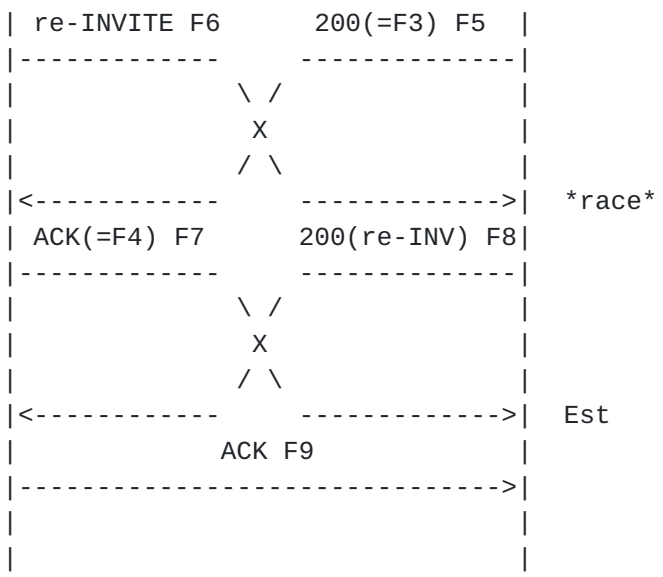
```
/* Alice transits to the Mortal state upon sending BYE.
   Therefore, after this, she does not begin a session even
   though she receives a 200 response with an answer.  */
```

F5 ACK Alice -> Bob

F6 200 OK (BYE) Bob -> Alice

3.1.4 Receiving re-INVITE (Established state) in Moratorium state (case 1)

```
  State  Alice                          Bob  State
         |                               |
         |         ini-INVITE F1         |
    Pre  |------------------------------>|  Pre
         |             180 F2            |
    Ear  |<------------------------------|  Ear
         |                               |
         |         200(ini-INV) F3       |
   Mora  |<------------------------------|  Mora
         |       ACK F4(packet loss)     |
    Est  |-------------------->x         |
```



This scenario illustrates the race condition which occurs when UAS receives re-INVITE request sent from the Established state, in the Moratorium state.
UAS receives a re-INVITE before receiving an ACK for ini-INVITE. UAS sends a 200 OK to the re-INVITE (F8) because it has sent a 200 OK to the ini-INVITE (F3, F5) and the dialog has already been established. (Because F5 is a retransmission of F3, SDP negotiation is not performed here.) If a 200 OK to the ini-INVITE has an offer and the answer is in the ACK, UA should return by a 491 to the re-INVITE (refer to 3.1.5). As it can be seen in Section 3.3.2 below, the 491 response seems to be closely related to session establishment, even in cases other than INVITE cross-over. This example recommends 200 be sent instead of 491 because it does not have influence on session. However, a 491 response can also lead to the same outcome, so the either response can be used.
Moreover, if UAS doesn't receive an ACK for a long time, it should send a BYE and terminate the dialog.

Message Details

F1 INVITE Alice -> Bob

```
INVITE sip:bob@biloxi.example.com SIP/2.0
Via: SIP/2.0/UDP client.atlanta.example.com:5060;branch=z9hG4bK74bf9
Max-Forwards: 70
From: Alice <sip:alice@atlanta.example.com>;tag=9fxced76sl
To: Bob <sip:bob@biloxi.example.com>
Call-ID: 3848276298220188511@atlanta.example.com
CSeq: 1 INVITE
```

```
   Contact: <sip:alice@client.atlanta.example.com;transport=udp>
   Content-Type: application/sdp
```

```
   Content-Length: 137

   v=0
   o=alice 2890844526 2890844526 IN IP4 client.atlanta.example.com
   s=-
   c=IN IP4 192.0.2.101
   t=0 0
   m=audio 49172 RTP/AVP 0
   a=rtpmap:0 PCMU/8000

   /* ini-INVITE contains an offer.  */


   F2 180 Ringing Bob -> Alice

   SIP/2.0 180 Ringing
   Via: SIP/2.0/UDP client.atlanta.example.com:5060;branch=z9hG4bK74bf9
    ;received=192.0.2.101
   From: Alice <sip:alice@atlanta.example.com>;tag=9fxced76sl
   To: Bob <sip:bob@biloxi.example.com>;tag=8321234356

   Call-ID: 3848276298220188511@atlanta.example.com
   CSeq: 1 INVITE
   Contact: <sip:bob@client.biloxi.example.com;transport=udp>
   Content-Length: 0


   F3 200 OK Bob -> Alice

   SIP/2.0 200 OK
   Via: SIP/2.0/UDP client.atlanta.example.com:5060;branch=z9hG4bK74bf9
    ;received=192.0.2.101
   From: Alice <sip:alice@atlanta.example.com>;tag=9fxced76sl
   To: Bob <sip:bob@biloxi.example.com>;tag=8321234356
   Call-ID: 3848276298220188511@atlanta.example.com
   CSeq: 1 INVITE
   Contact: <sip:bob@client.biloxi.example.com;transport=udp>
   Content-Type: application/sdp
   Content-Length: 133

   v=0
   o=bob 2890844527 2890844527 IN IP4 client.biloxi.example.com
   s=-
   c=IN IP4 192.0.2.201
   t=0 0
   m=audio 3456 RTP/AVP 0
   a=rtpmap:0 PCMU/8000
```

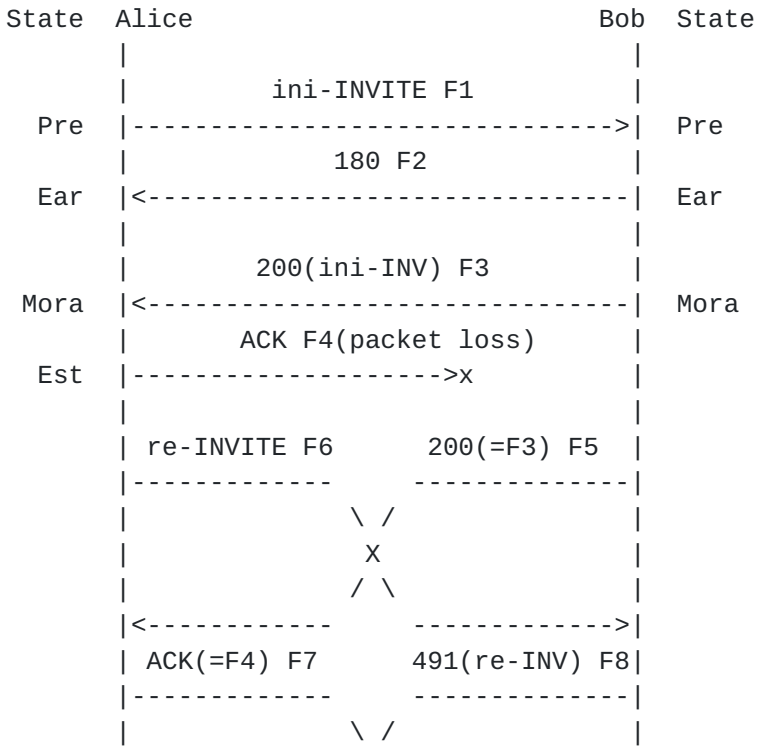
F4 ACK Alice -> Bob

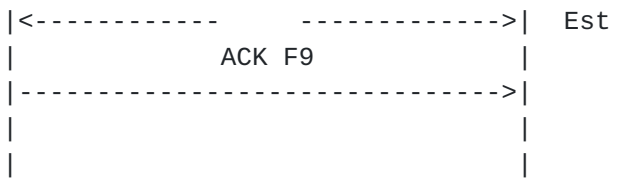
```
   ACK sip:bob@client.biloxi.example.com SIP/2.0
   Via: SIP/2.0/UDP client.atlanta.example.com:5060;branch=z9hG4bKnashds8
   Max-Forwards: 70
   From: Alice <sip:alice@atlanta.example.com>;tag=9fxced76sl
   To: Bob <sip:bob@biloxi.example.com>;tag=8321234356
   Call-ID: 3848276298220188511@atlanta.example.com
   CSeq: 1 ACK
   Content-Length: 0

   /* ACK request is lost.  */


   F5 200 OK (=F3) Bob -> Alice (retransmission)
   /* UAS retransmits a 200 OK to the ini-INVITE since it has not
      received an ACK.  */


   F6 re-INVITE Alice -> Bob

   INVITE sip:sip:bob@client.biloxi.example.com SIP/2.0
   Via: SIP/2.0/UDP client.atlanta.example.com:5060;branch=z9hG4bK74bf9.1
   Max-Forwards: 70
   From: Alice <sip:alice@atlanta.example.com>;tag=9fxced76sl
   To: Bob <sip:bob@biloxi.example.com>;tag=8321234356
   Call-ID: 3848276298220188511@atlanta.example.com
   CSeq: 2 INVITE
   Content-Length: 147

   v=0
   o=alice 2890844526 2890844527 IN IP4 client.atlanta.example.com
   s=-
   c=IN IP4 192.0.2.101
   t=0 0
   m=audio 49172 RTP/AVP 0
   a=rtpmap:0 PCMU/8000
   a=sendonly


   F7 ACK (=F4) Alice -> Bob (retransmission)

   F8 200 OK (re-INVITE) Bob -> Alice

   SIP/2.0 200 OK
   Via: SIP/2.0/UDP client.atlanta.example.com:5060;branch=z9hG4bK74bf9.1
   Max-Forwards: 70
   From: Alice <sip:alice@atlanta.example.com>;tag=9fxced76sl
   To: Bob <sip:bob@biloxi.example.com>;tag=8321234356
   Call-ID: 3848276298220188511@atlanta.example.com
```

```
   CSeq: 2 INVITE
```

```
Content-Length: 143

v=0
o=bob 2890844527 2890844528 IN IP4 client.biloxi.example.com
s=-
c=IN IP4 192.0.2.201
t=0 0
m=audio 3456 RTP/AVP 0
a=rtpmap:0 PCMU/8000
a=recvonly


F9 ACK Alice -> Bob

ACK sip:sip:bob@client.biloxi.example.com SIP/2.0
Via: SIP/2.0/UDP client.atlanta.example.com:5060;branch=z9hG4bK230f2.1
Max-Forwards: 70
From: Alice <sip:alice@atlanta.example.com>;tag=9fxced76sl
To: Bob <sip:bob@biloxi.example.com>;tag=8321234356
Call-ID: 3848276298220188511@atlanta.example.com
CSeq: 2 ACK
Content-Length: 0
```

3.1.5 Receiving re-INVITE (Established state) in Moratorium state (case 2)





This scenario is basically the same as that of Section 3.1.4, but differs in sending an offer in 200 and an answer in ACK. Different to the previous case, the offer in the 200 (F3) and the offer in the re-INVITE (F6) collide with each other.
Bob sends 491 to re-INVITE since he is not able to properly handle a new request until he receives an answer.

Message Details

F1 INVITE Alice -> Bob

```
INVITE sip:bob@biloxi.example.com SIP/2.0
Via: SIP/2.0/UDP client.atlanta.example.com:5060;branch=z9hG4bK74bf9
Max-Forwards: 70
From: Alice <sip:alice@atlanta.example.com>;tag=9fxced76sl
To: Bob <sip:bob@biloxi.example.com>
Call-ID: 3848276298220188511@atlanta.example.com
CSeq: 1 INVITE
Contact: <sip:alice@client.atlanta.example.com;transport=udp>
Content-Length: 0

/* The request does not contain an offer.  */
```

F2 180 Ringing Bob -> Alice

F3 200 OK Bob -> Alice

```
SIP/2.0 200 OK
Via: SIP/2.0/UDP client.atlanta.example.com:5060;branch=z9hG4bK74bf9
 ;received=192.0.2.101
From: Alice <sip:alice@atlanta.example.com>;tag=9fxced76sl
To: Bob <sip:bob@biloxi.example.com>;tag=8321234356
Call-ID: 3848276298220188511@atlanta.example.com
CSeq: 1 INVITE
Contact: <sip:bob@client.biloxi.example.com;transport=udp>
Content-Type: application/sdp
Content-Length: 133

v=0
o=bob 2890844527 2890844527 IN IP4 client.biloxi.example.com
```

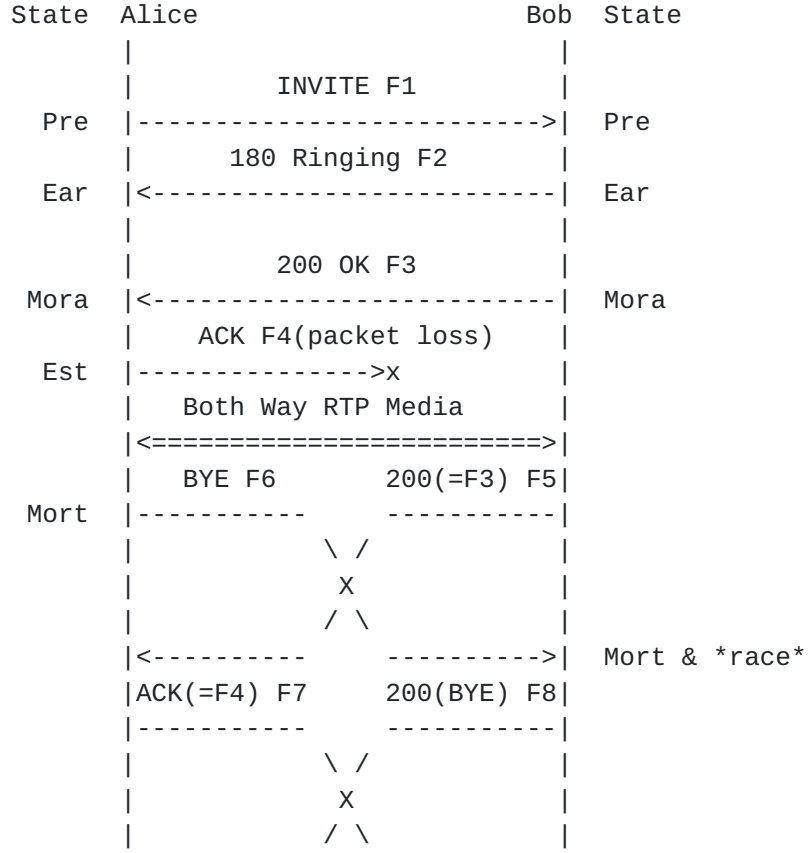
```
s=-
```

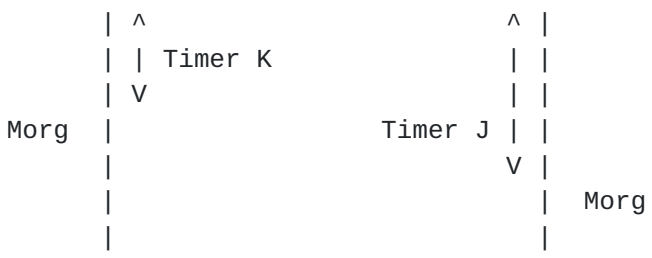
```
   c=IN IP4 192.0.2.201
   t=0 0
   m=audio 3456 RTP/AVP 0
   a=rtpmap:0 PCMU/8000

   /* An offer is made in 200 */


   F4 ACK Alice -> Bob

   ACK sip:bob@client.biloxi.example.com SIP/2.0
   Via: SIP/2.0/UDP client.atlanta.example.com:5060;branch=z9hG4bKnashds8
   Max-Forwards: 70
   From: Alice <sip:alice@atlanta.example.com>;tag=9fxced76sl
   To: Bob <sip:bob@biloxi.example.com>;tag=8321234356
   Call-ID: 3848276298220188511@atlanta.example.com
   CSeq: 1 ACK
   Content-Type: application/sdp
   Content-Length: 137

   v=0
   o=alice 2890844526 2890844526 IN IP4 client.atlanta.example.com
   s=-
   c=IN IP4 192.0.2.101
   t=0 0
   m=audio 49172 RTP/AVP 0
   a=rtpmap:0 PCMU/8000

   /* The request contains an answer, but the request is lost.  */


   F5 200 OK (=F3) Bob -> Alice (retransmission)
   /* UAS retransmits a 200 OK to the ini-INVITE since it has not
      received an ACK.  */


   F6 re-INVITE Alice -> Bob

   INVITE sip:sip:bob@client.biloxi.example.com SIP/2.0

   Via: SIP/2.0/UDP client.atlanta.example.com:5060;branch=z9hG4bK74bf9.1
   Max-Forwards: 70
   From: Alice <sip:alice@atlanta.example.com>;tag=9fxced76sl
   To: Bob <sip:bob@biloxi.example.com>;tag=8321234356
   Call-ID: 3848276298220188511@atlanta.example.com
   CSeq: 2 INVITE
   Content-Length: 147

   v=0
```

```
   o=alice 2890844526 2890844527 IN IP4 client.atlanta.example.com
```

```
   s=-
   c=IN IP4 192.0.2.101
   t=0 0
   m=audio 49172 RTP/AVP 0
   a=rtpmap:0 PCMU/8000
   a=sendonly

   /* The request contains an offer.  */


   F7 ACK (=F4) Alice -> Bob (retransmission)
   /* A retransmission triggered by the reception of a retransmitted
      200.  */


   F8 491 (re-INVITE) Bob -> Alice
   /* Bob sends 491 (Request Pending), since Bob has a pending
      offer.  */

   F9 ACK Alice -> Bob
```

3.1.6 Receiving BYE (Established state) in Moratorium state





This scenario illustrates the race condition which occurs when UAS receives an Established message, BYE, in the Moratorium state.
An ACK request for a 200 OK response is lost (or delayed).
Immediately after Bob retransmits the 200 OK to ini-INVITE, and Alice sends a BYE request at the same time.
Depending on the implementation of a SIP UA, Alice may want to start a session again by the reception of the retransmitted 200 OK with SDP since she has already terminated a session by sending a BYE. In that case, when UAC receives a retransmitted 200 OK after sending a BYE, a session should not be started again since the session which is not associated with dialog still remains. Moreover, in the case where UAS sends an offer in a 200 OK , UAS should not start a session again for the same reason if UAS receives a retransmitted ACK after receiving a BYE.

Message Details

F1 INVITE Alice -> Bob

F2 180 Ringing Bob -> Alice

F3 200 OK Bob -> Alice

F4 ACK Alice -> Bob

```
ACK sip:bob@client.biloxi.example.com SIP/2.0
Via: SIP/2.0/UDP client.atlanta.example.com:5060;branch=z9hG4bKnashds8
Max-Forwards: 70
From: Alice <sip:alice@atlanta.example.com>;tag=9fxced76sl
To: Bob <sip:bob@biloxi.example.com>;tag=8321234356
Call-ID: 3848276298220188511@atlanta.example.com
CSeq: 1 ACK
Content-Length: 0
```

/* ACK request is lost. */

F5 200 OK (retransmission) Bob -> Alice

```
   SIP/2.0 200 OK
```

```
   Via: SIP/2.0/UDP client.atlanta.example.com:5060;branch=z9hG4bK74bf9
    ;received=192.0.2.101
   From: Alice <sip:alice@atlanta.example.com>;tag=9fxced76sl
   To: Bob <sip:bob@biloxi.example.com>;tag=8321234356
   Call-ID: 3848276298220188511@atlanta.example.com
   CSeq: 1 INVITE
   Contact: <sip:bob@client.biloxi.example.com;transport=udp>
   Content-Type: application/sdp
   Content-Length: 133

   v=0
   o=bob 2890844527 2890844527 IN IP4 client.biloxi.example.com
   s=-
   c=IN IP4 192.0.2.201
   t=0 0
   m=audio 3456 RTP/AVP 0
   a=rtpmap:0 PCMU/8000

   /* UAS retransmits a 200 OK to the ini-INVITE since it has not
      received an ACK.  */


   F6 BYE Alice -> Bob

   BYE sip:bob@client.biloxi.example.com SIP/2.0
   Via: SIP/2.0/UDP client.atlanta.example.com:5060;branch=z9hG4bKnashds9
   Max-Forwards: 70
   From: Alice <sip:alice@atlanta.example.com>;tag=9fxced76sl
   To: Bob <sip:bob@biloxi.example.com>;tag=8321234356
   Call-ID: 3848276298220188511@atlanta.example.com
   CSeq: 2 BYE
   Content-Length: 0

   /* Bob retransmits a 200 OK and Alice sends a BYE at the same time.
      Alice transits to the Mortal state, so she does not begin a
      session after this even though she receives a 200 response to
      the re-INVITE.  */

   F7 ACK(=F4) Alice -> Bob

   F8 200 OK (BYE) Bob -> Alice

   SIP/2.0 200 OK
   Via: SIP/2.0/UDP client.atlanta.example.com:5060;branch=z9hG4bKnashds9
    ;received=192.0.2.101
   From: Alice <sip:alice@atlanta.example.com>;tag=9fxced76sl
   To: Bob <sip:bob@biloxi.example.com>;tag=8321234356
   Call-ID: 3848276298220188511@atlanta.example.com
   CSeq: 2 BYE
```

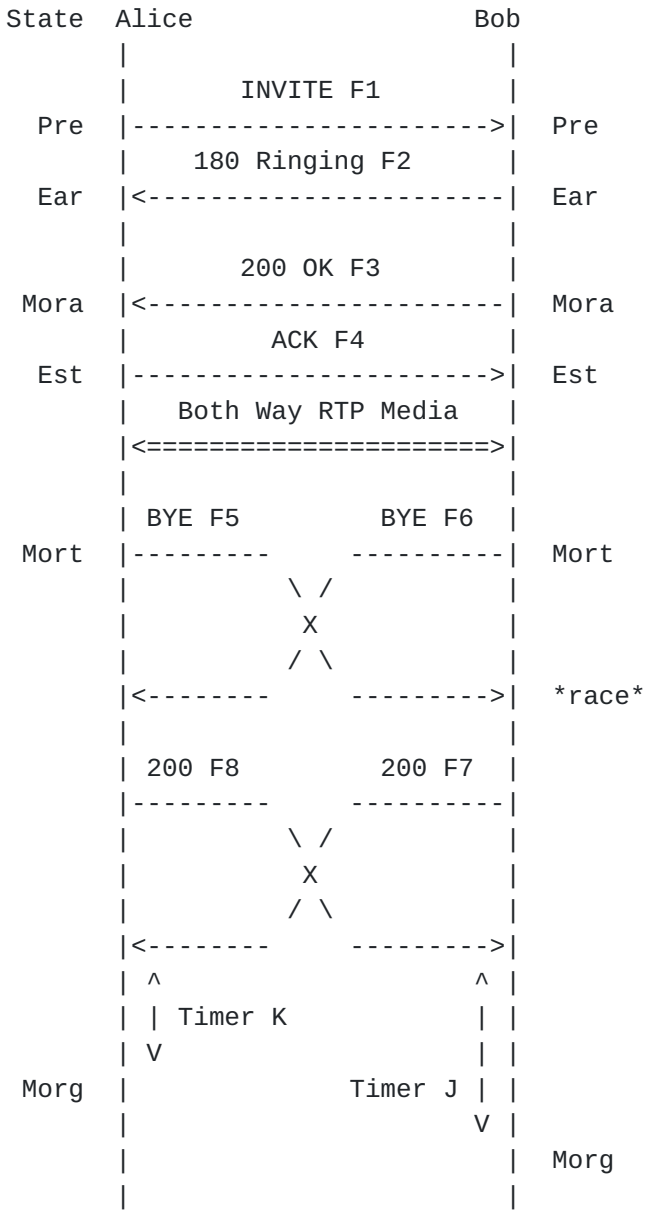
```
   Content-Length: 0
```

/* Bob sends a 200 OK to the BYE. */

3.2 Receiving message in the Mortal State

This section shows some examples of call flow in race condition when receiving the message from other states in the Mortal state.

3.2.1 Receiving BYE (Establish state) in Mortal state



This scenario illustrates the race condition which occurs when UAS

receives an Established message, BYE, in the Mortal state.
Alice and Bob send a BYE at the same time. A dialog and session is ended shortly after a BYE request is passed to a client transaction. As shown in Section 2, UA remains in the Mortal state.
UAs in the Mortal state return error responses to the requests that operate dialog or session, such as re-INVITE, UPDATE, or REFER. However, UA shall return 200 OK to the BYE taking the use case into consideration where BYE request is sent by both a caller and a callee to exchange reports about the session when it is being terminated. (Since the dialogue and the session both terminate when a BYE is sent, the choice of sending 200 or an error response upon receiving BYE in the Mortal state does not affect the resulting termination. Therefore, even though this example uses a 200 response, other responses can also be used.)

Message Details

F1 INVITE Alice -> Bob

F2 180 Ringing Bob -> Alice

F3 200 OK Bob -> Alice

F4 ACK Alice -> Bob

F5 BYE Alice -> Bob

```
BYE sip:bob@client.biloxi.example.com SIP/2.0
Via: SIP/2.0/UDP client.atlanta.example.com:5060;branch=z9hG4bKnashds8
Max-Forwards: 70
From: Alice <sip:alice@atlanta.example.com>;tag=9fxced76sl
To: Bob <sip:bob@biloxi.example.com>;tag=8321234356
Call-ID: 3848276298220188511@atlanta.example.com
CSeq: 2 BYE
Content-Length: 0
```

```
/* The session is terminated at the moment Alice sends a BYE.
   The dialog still exists then, but it is certain to be terminated
   in a short period of time.  The dialog is completely
   terminated when the timeout of the BYE request occurs.  */
```

F6 BYE Bob -> Alice

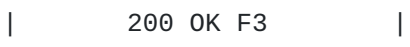
```
BYE sip:alice@client.atlanta.example.com SIP/2.0
Via: SIP/2.0/UDP client.biloxi.example.com:5060;branch=z9hG4bKnashds7
Max-Forwards: 70
From: Bob <sip:bob@biloxi.example.com>;tag=8321234356
```

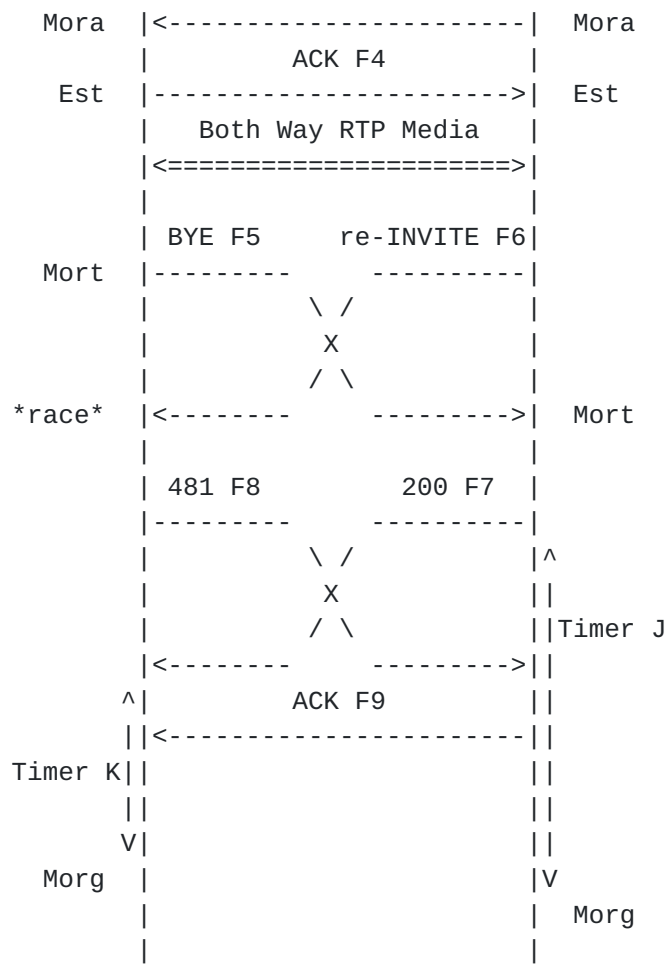
```
   To: Alice <sip:alice@atlanta.example.com>;tag=9fxced76sl
```

```
   Call-ID: 3848276298220188511@atlanta.example.com
   CSeq: 1 BYE
   Content-Length: 0

   /* Bob has also transmitted a BYE simultaneously with Alice.
      Bob terminates the session and the dialog.  */


   F7 200 OK Bob -> Alice

   SIP/2.0 200 OK
   Via: SIP/2.0/UDP client.atlanta.example.com:5060;branch=z9hG4bKnashds8
    ;received=192.0.2.201
   From: Alice <sip:alice@atlanta.example.com>;tag=9fxced76sl
   To: Bob <sip:bob@biloxi.example.com>;tag=8321234356
   Call-ID: 3848276298220188511@atlanta.example.com
   CSeq: 2 BYE
   Content-Length: 0

   /* Since the dialog is in the Moratorium state, Bob responds with
      a 200 to the BYE request.  */


   F8 200 OK Alice -> Bob

   SIP/2.0 200 OK
   Via: SIP/2.0/UDP client.biloxi.example.com:5060;branch=z9hG4bKnashds7
    ;received=192.0.2.201
   From: Bob <sip:bob@biloxi.example.com>;tag=8321234356
   To: Alice <sip:alice@atlanta.example.com>;tag=9fxced76sl
   Call-ID: 3848276298220188511@atlanta.example.com
   CSeq: 1 BYE
   Content-Length: 0

   /* Since Alice has transited from the Established state to the Mortal
      state by sending BYE, Alice responds with a 200 to the BYE
      request.  */
```

3.2.2 Receiving re-INVITE (Establish state) in Mortal state

```
  State  Alice                  Bob
         |                       |
         |       INVITE F1       |
    Pre  |---------------------->|  Pre
         |    180 Ringing F2     |
    Ear  |<----------------------|  Ear
         |                       |
```


200 OK F3



This scenario illustrates the race condition which occurs when UAS receives an Established message, re-INVITE, in the Mortal state. Bob sends a re-INVITE, and Alice sends a BYE at the same time. The re-INVITE is responded by a 481, since the TU of Alice has transited from the Established state to the Mortal state by sending BYE. Bob sends ACK for the 481 response, because the ACK for error responses is handled by the transaction layer and at the point of receiving the 481 the INVITE client transaction still remains (even though the dialog has been terminated).

Message Details

F1 INVITE Alice -> Bob

F2 180 Ringing Bob -> Alice

F3 200 OK Bob -> Alice

F4 ACK Alice -> Bob

F5 BYE Alice -> Bob

```
BYE sip:bob@client.biloxi.example.com SIP/2.0
Via: SIP/2.0/UDP client.atlanta.example.com:5060;branch=z9hG4bKnashds8
Max-Forwards: 70
From: Alice <sip:alice@atlanta.example.com>;tag=9fxced76sl
To: Bob <sip:bob@biloxi.example.com>;tag=8321234356
Call-ID: 3848276298220188511@atlanta.example.com
CSeq: 2 BYE
Content-Length: 0
```

```
/* Alice sends a BYE and terminates the session, and transits from
   the Established state to the Mortal state.  */
```

F6 re-INVITE Bob -> Alice

```
INVITE sip:alice@client.atlanta.example.com SIP/2.0
Via: SIP/2.0/UDP client.biloxi.example.com:5060;branch=z9hG4bKnashds7
Session-Expires: 300;refresher=uac
Supported: timer
Max-Forwards: 70
From: Bob <sip:bob@biloxi.example.com>;tag=8321234356
To: Alice <sip:alice@atlanta.example.com>;tag=9fxced76sl
Call-ID: 3848276298220188511@atlanta.example.com
CSeq: 1 INVITE
Content-Length: 0
```

```
/* Alice sends a BYE, and Bob sends a re-INVITE at the same time.
   The dialog state transits to the Mortal state at the moment
   Alice sends the BYE, but Bob does not know it until he receives
   the BYE.  Therefore, the dialog is in the Terminated state from
   Alice's point of view, but it is the Confirmed state
   from Bob's point of view.  A race condition occurs.  */
```

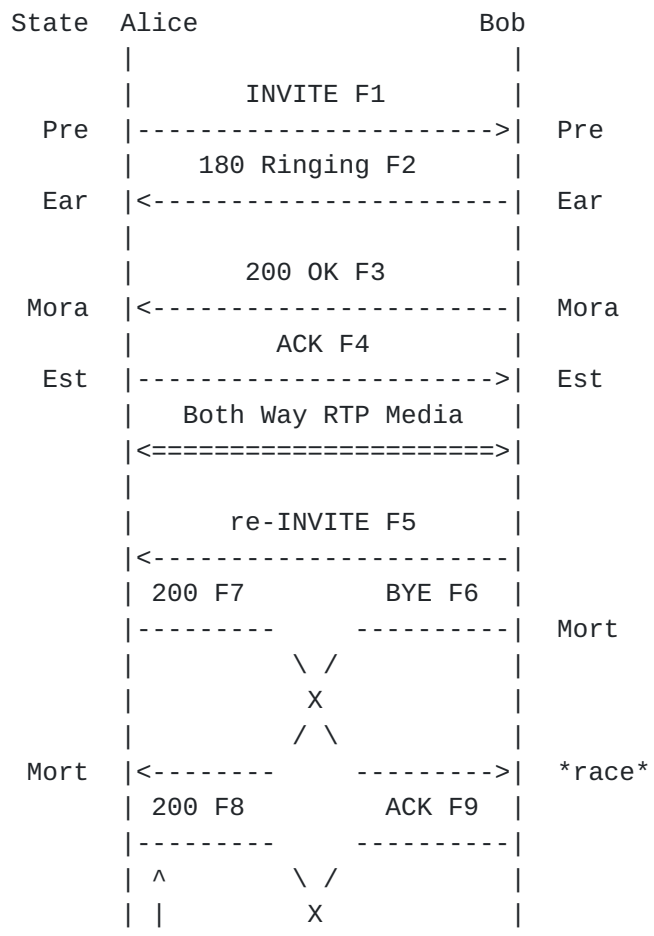
F7 200 OK Bob -> Alice

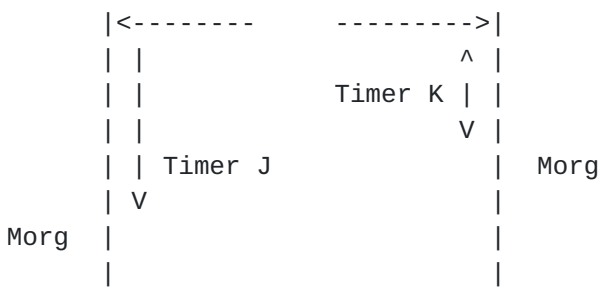
```
SIP/2.0 200 OK
Via: SIP/2.0/UDP client.atlanta.example.com:5060;branch=z9hG4bKnashds8
 ;received=192.0.2.201
From: Alice <sip:alice@atlanta.example.com>;tag=9fxced76sl
To: Bob <sip:bob@biloxi.example.com>;tag=8321234356
Call-ID: 3848276298220188511@atlanta.example.com

CSeq: 2 BYE
Content-Length: 0
```

F8 481 Call/Transaction Does Not Exist Alice -> Bob

```
SIP/2.0 481 Call/Transaction Does Not Exist
Via: SIP/2.0/UDP client.biloxi.example.com:5060;branch=z9hG4bKnashds7
 ;received=192.0.2.201
From: Bob <sip:bob@biloxi.example.com>;tag=8321234356
To: Alice <sip:alice@atlanta.example.com>;tag=9fxced76sl
Call-ID: 3848276298220188511@atlanta.example.com
CSeq: 1 INVITE
Content-Length: 0

/* Since Alice is in the Mortal state, she responds with a 481 to the
   re-INVITE.  */


F9 ACK Bob -> Alice

/* ACK for an error response is handled by Bob's INVITE client
   transaction. */
```

3.2.3 Receiving 200OK for re-INVITE (Established state) in Mortal state





This scenario illustrates the race condition which occurs when UAS receives an Established message, 200 to re-INVITE, in the Mortal state. Bob sends a BYE immediately after sending a re-INVITE. (A user is not conscious that refresher sends re-INVITE automatically. For example, in the case of a telephone application, it is possible that a user places a receiver immediately after refresher.)
Bob sends ACK for a 200 response to INVITE in the Mortal state, so that he completes the INVITE transaction.

Message Details

F1 INVITE Alice -> Bob

F2 180 Ringing Bob -> Alice

F3 200 OK Bob -> Alice

F4 ACK Alice -> Bob

F5 re-INVITE Bob -> Alice

```
INVITE sip:alice@client.atlanta.example.com SIP/2.0
Via: SIP/2.0/UDP client.biloxi.example.com:5060;branch=z9hG4bKnashds7
Session-Expires: 300;refresher=uac
Supported: timer
Max-Forwards: 70
From: Bob <sip:bob@biloxi.example.com>;tag=8321234356
To: Alice <sip:alice@atlanta.example.com>;tag=9fxced76sl
Call-ID: 3848276298220188511@atlanta.example.com
CSeq: 1 INVITE
Content-Length: 0
```

F6 BYE Bob -> Alice

```
BYE sip:alice@client.atlanta.example.com SIP/2.0
Via: SIP/2.0/UDP client.biloxi.example.com:5060;branch=z9hG4bKnashds8
```

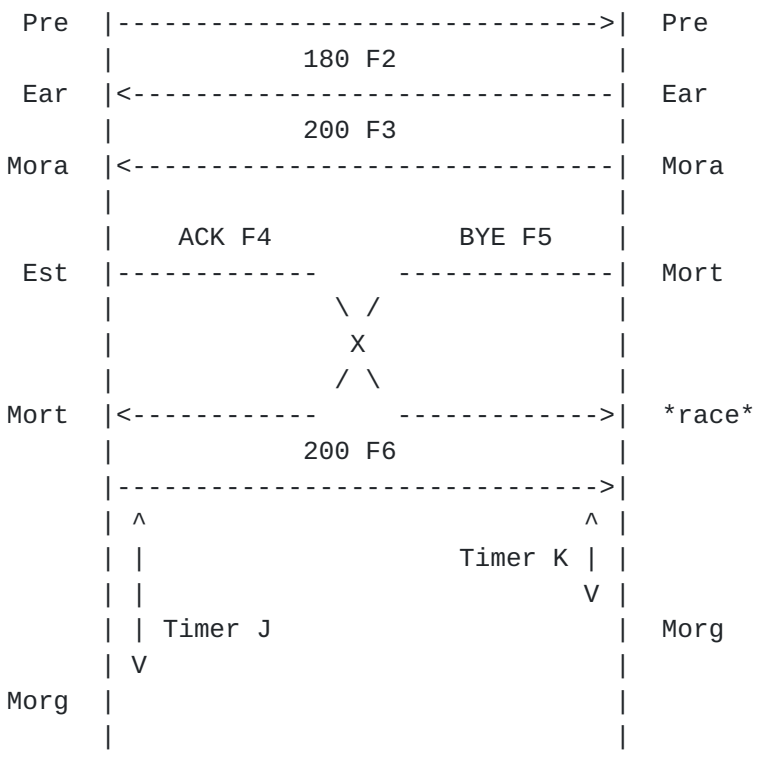
```
   Max-Forwards: 70
```

```
   From: Bob <sip:bob@biloxi.example.com>;tag=8321234356
   To: Alice <sip:alice@atlanta.example.com>;tag=9fxced76sl
   Call-ID: 3848276298220188511@atlanta.example.com
   CSeq: 2 BYE
   Content-Length: 0

   /* Bob sends BYE immediately after sending the re-INVITE.
      Bob terminates the session and transits from the Established
      state to the Mortal state.  */


   F7 200 OK (re-INVITE) Alice -> Bob

   SIP/2.0 200 OK
   Via: SIP/2.0/UDP client.atlanta.example.com:5060;branch=z9hG4bKnashds7
    ;received=192.0.2.201
   From: Bob <sip:bob@biloxi.example.com>;tag=8321234356
   To: Alice <sip:alice@atlanta.example.com>;tag=9fxced76sl
   Call-ID: 3848276298220188511@atlanta.example.com
   CSeq: 1 INVITE
   Content-Length: 0

   /* Bob sends BYE, and Alice responds with a 200 OK to the re-INVITE.
      A race condition occurs.  */


   F8 200 OK (BYE) Alice -> Bob

   SIP/2.0 200 OK
   Via: SIP/2.0/UDP client.biloxi.example.com:5060;branch=z9hG4bKnashds8
    ;received=192.0.2.201
   From: Bob <sip:bob@biloxi.example.com>;tag=8321234356
   To: Alice <sip:alice@atlanta.example.com>;tag=9fxced76sl
   Call-ID: 3848276298220188511@atlanta.example.com
   CSeq: 2 BYE
   Content-Length: 0


   F9 ACK Bob -> Alice

   /* Bob sends ACK in the Mortal state to complete the three-way
      handshake of the INVITE transaction.  */
```

3.2.4 Receiving ACK (Moratorium state) in Mortal state

```
  State  Alice                          Bob  State
         |                               |
```


ini-INVITE F1



This scenario illustrates the race condition which occurs when UAS receives an Established message, ACK to 200, in the Mortal state. Alice sends an ACK and Bob sends a BYE at the same time. When the offer is in a 2xx, and the answer is in an ACK, this example is in a race condition. The session is not started by receiving the ACK because Bob has already terminated the session by sending the BYE. The answer in the ACK request is just ignored.

F1 INVITE Alice -> Bob

F2 180 Ringing Bob -> Alice

F3 200 OK Bob -> Alice

F4 ACK Alice -> Bob

```
ACK sip:bob@client.biloxi.example.com SIP/2.0
Via: SIP/2.0/UDP client.atlanta.example.com:5060;branch=z9hG4bK74bd5
Max-Forwards: 70
From: Alice <sip:alice@atlanta.example.com>;tag=9fxced76sl
To: Bob <sip:bob@biloxi.example.com>;tag=8321234356
Call-ID: 3848276298220188511@atlanta.example.com
CSeq: 1 ACK
Content-Length: 0
```

```
   /* RTP streams are established between Alice and Bob */
```

F5 BYE Alice -> Bob

```
BYE sip:bob@client.biloxi.example.com SIP/2.0
Via: SIP/2.0/UDP client.atlanta.example.com:5060;branch=z9hG4bKnashds8
Max-Forwards: 70
From: Alice <sip:alice@atlanta.example.com>;tag=9fxced76sl
To: Bob <sip:bob@biloxi.example.com>;tag=8321234356
Call-ID: 3848276298220188511@atlanta.example.com
CSeq: 2 BYE
Content-Length: 0
```

/* Alice sends a BYE and terminates the session and dialog. */

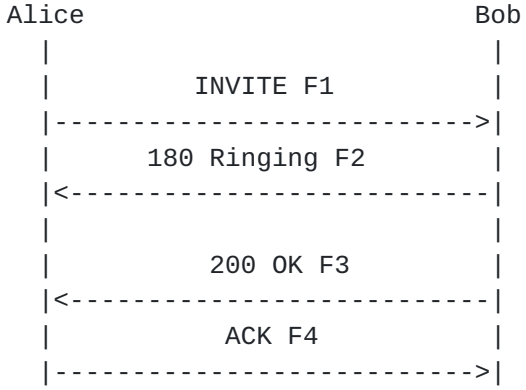
F6 200 OK Bob -> Alice

```
SIP/2.0 200 OK
Via: SIP/2.0/UDP client.atlanta.example.com:5060;branch=z9hG4bKnashds8
 ;received=192.0.2.201
From: Alice <sip:alice@atlanta.example.com>;tag=9fxced76sl
To: Bob <sip:bob@biloxi.example.com>;tag=8321234356
Call-ID: 3848276298220188511@atlanta.example.com
CSeq: 2 BYE
Content-Length: 0
```

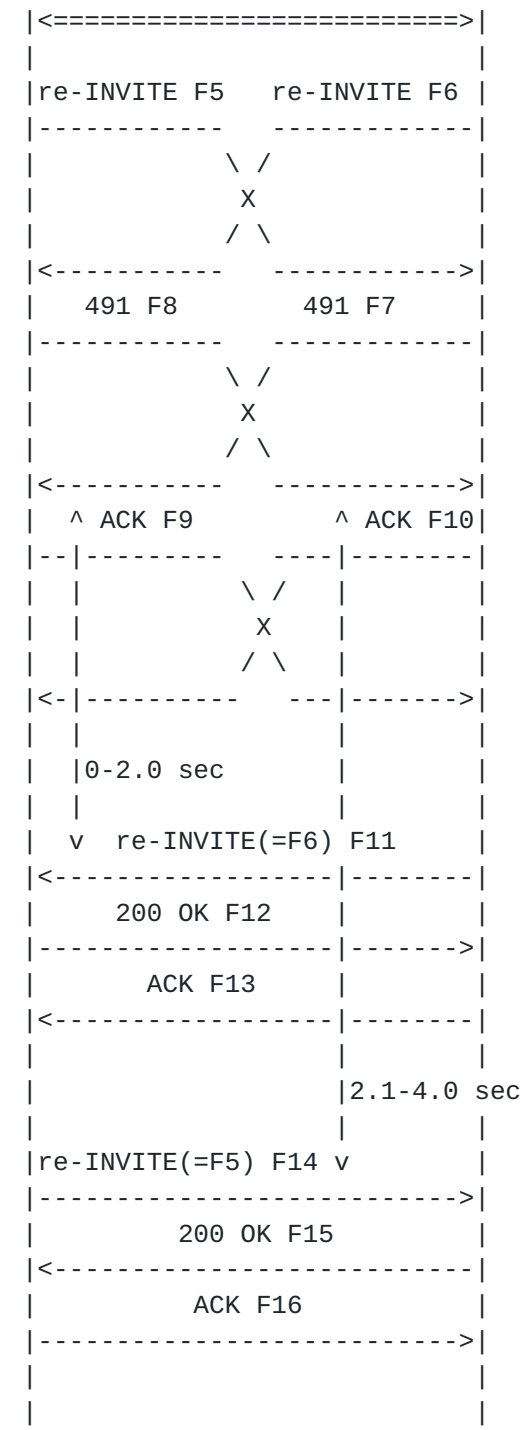
3.3 Other race condition

This section shows the examples in race condition that are not directly related to the dialog state transition. Here explains the way to handle the race condition generated when UAs treat "What is established by SIP", which has some relation to dialog.

3.3.1 re-INVITE crossover




Both Way RTP Media



In this scenario, Alice and Bob send re-INVITE at the same time. When two re-INVITEs cross in the same dialog, they resend re-INVITEs after different interval for each, according to Section 14.1, of RFC3261 [1]. When Alice sends the re-INVITE and it crosses, the re-INVITE will be sent again after 2.1-4.0 seconds because she owns the Call-ID (she generated it). Bob will send an INVITE again after

0.0-2.0 seconds, because Bob isn't the owner of the Call-ID.
Therefore, each user agent must remember whether it has generated the

Call-ID of the dialog or not, in case an INVITE may cross with another INVITE.
In this example, Alice's re-INVITE is for session modification and Bob's re-INVITE is for session refresh. In this case, after the 491 responses, Bob retransmits the re-INVITE for session refresh earlier than Alice. If Alice was to retransmit her re-INVITE (that is, if she was not the owner of Call-ID), the request would refresh and modify the session at the same time. Then Bob would know that he would not need to retransmit his re-INVITE to refresh the session.
In another instance where two re-INVITEs for session modification, cross over, retransmitting the same re-INVITE again after 491 by the Call-ID owner (the UA which retransmits its re-INVITE after the other UA) may result in a behavior different from what the user originally intended to, so the UA needs to decide if the retransmission of the re-INVITE is necessary.
(For example, when a call hold and an addition of video media cross over, mere retransmission of the re-INVITE at the firing of the timer may result in the situation where the video is transmitted immediately after the holding of the audio. This behavior is probably not intended by the users.)

Message Details

F1 INVITE Alice -> Bob

F2 180 Ringing Bob -> Alice

F3 200 OK Bob -> Alice

F4 ACK Alice -> Bob

F5 re-INVITE Alice -> Bob

```
INVITE sip:sip:bob@client.biloxi.example.com SIP/2.0
Via: SIP/2.0/UDP client.atlanta.example.com:5060;branch=z9hG4bK74bf9
Max-Forwards: 70
From: Alice <sip:alice@atlanta.example.com>;tag=9fxced76sl
To: Bob <sip:bob@biloxi.example.com>;tag=8321234356
Call-ID: 3848276298220188511@atlanta.example.com
CSeq: 2 INVITE
Content-Length: 147

v=0
o=alice 2890844526 2890844527 IN IP4 client.atlanta.example.com
s=-
c=IN IP4 192.0.2.101
t=0 0
m=audio 49172 RTP/AVP 0
```

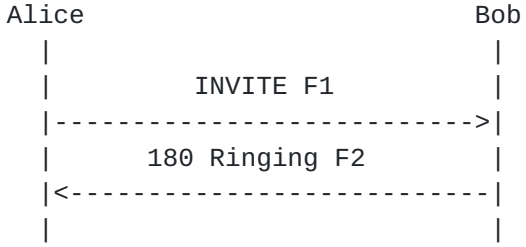
```
   a=rtpmap:0 PCMU/8000
```

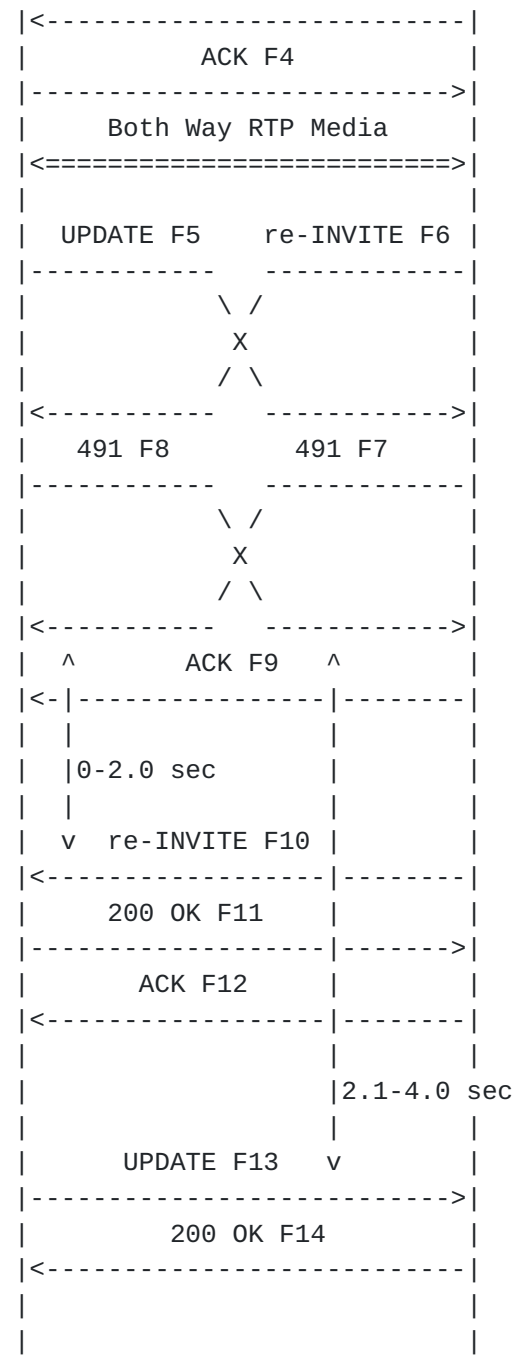
```
   a=sendonly

   /* re-INVITE for session modification (a=sendrecv -> sendonly).  */


   F6 re-INVITE Bob -> Alice

   INVITE sip:alice@client.atlanta.example.com SIP/2.0
   Via: SIP/2.0/UDP client.biloxi.example.com:5060;branch=z9hG4bKnashds7
   Session-Expires: 300;refresher=uac
   Supported: timer
   Max-Forwards: 70
   From: Bob <sip:bob@biloxi.example.com>;tag=8321234356
   To: Alice <sip:alice@atlanta.example.com>;tag=9fxced76sl
   Call-ID: 3848276298220188511@atlanta.example.com
   CSeq: 1 INVITE
   Content-Length: 0

   /* A re-INVITE request for a session refresh and that for
      a call hold are sent at the same time.  */


   F7 491 Request Pending Bob -> Alice
   /* Since a re-INVITE is in progress, a 491 response is returned.  */

   F8 491 Request Pending Alice -> Bob

   F9 ACK (INVITE) Alice -> Bob

   F10 ACK (INVITE) Bob -> Alice

   F11 re-INVITE Bob -> Alice

   INVITE sip:alice@client.atlanta.example.com SIP/2.0
   Via: SIP/2.0/UDP client.biloxi.example.com:5060;branch=z9hG4bKnashds7.1
   Session-Expires: 300;refresher=uac
   Supported: timer
   Max-Forwards: 70
   From: Bob <sip:bob@biloxi.example.com>;tag=8321234356
   To: Alice <sip:alice@atlanta.example.com>;tag=9fxced76sl
   Call-ID: 3848276298220188511@atlanta.example.com
   CSeq: 2 INVITE
   Content-Type: application/sdp
   Content-Length: 133

   v=0
   o=bob 2890844527 2890844527 IN IP4 client.biloxi.example.com
   s=-
   c=IN IP4 192.0.2.201
```

```
   t=0 0
```

```
   m=audio 3456 RTP/AVP 0
   a=rtpmap:0 PCMU/8000

   /* Since Bob is not the owner of the Call-ID, he sends a re-INVITE
      again after 0.0-2.0 seconds.  */


   F12 200 OK Alice -> Bob

   F13 ACK Bob -> Alice

   F14 re-INVITE Alice -> Bob

   INVITE sip:sip:bob@client.biloxi.example.com SIP/2.0
   Via: SIP/2.0/UDP client.atlanta.example.com:5060;branch=z9hG4bK74bf9.1
   Max-Forwards: 70
   From: Alice <sip:alice@atlanta.example.com>;tag=9fxced76sl
   To: Bob <sip:bob@biloxi.example.com>;tag=8321234356
   Call-ID: 3848276298220188511@atlanta.example.com
   CSeq: 3 INVITE
   Content-Length: 147

   v=0
   o=alice 2890844526 2890844527 IN IP4 client.atlanta.example.com
   s=-
   c=IN IP4 192.0.2.101
   t=0 0
   m=audio 49172 RTP/AVP 0
   a=rtpmap:0 PCMU/8000
   a=sendonly

   /* Since Alice is the owner of the Call-ID, Alice sends a re-INVITE
      again after 2.1-4.0 seconds.  */


   F15 200 OK Bob -> Alice

   F16 ACK Alice -> Bob
```

3.3.2 UPDATE and re-INVITE crossover





In this scenario, the UPDATE contains a SDP offer, therefore the UPDATE and re-INVITE are responded with 491 as in the case of "re-INVITE crossover". When an UPDATE for refresher which doesn't contain a session description and a re-INVITE crossed each other, both requests succeed with 200 (491 means that UA have a pending request). Moreover, the same is equally true for UPDATE crossover. In the former case where either UPDATE contains a session description the requests fail with 491, and in the latter cases succeed with 200.

Editor's Note:

A 491 response is considered as a result that UA judged the effectiveness of request to "What is established by SIP". Therefore, it is considered that 491 will be used in all the requests that demand operation to "What is established by SIP".

Message Details

F1 INVITE Alice -> Bob

F2 180 Ringing Bob -> Alice

F3 200 OK Bob -> Alice

F4 ACK Alice -> Bob

F5 UPDATE Alice -> Bob

```
UPDATE sip:sip:bob@client.biloxi.example.com SIP/2.0
Via: SIP/2.0/UDP client.atlanta.example.com:5060;branch=z9hG4bK74bf9
Max-Forwards: 70
From: Alice <sip:alice@atlanta.example.com>;tag=9fxced76sl
To: Bob <sip:bob@biloxi.example.com>;tag=8321234356
Call-ID: 3848276298220188511@atlanta.example.com
CSeq: 2 UPDATE
Content-Length: 147

v=0
o=alice 2890844526 2890844527 IN IP4 client.atlanta.example.com
s=-
c=IN IP4 192.0.2.101
t=0 0
m=audio 49172 RTP/AVP 0
a=rtpmap:0 PCMU/8000
a=sendonly
```

F6 re-INVITE Bob -> Alice

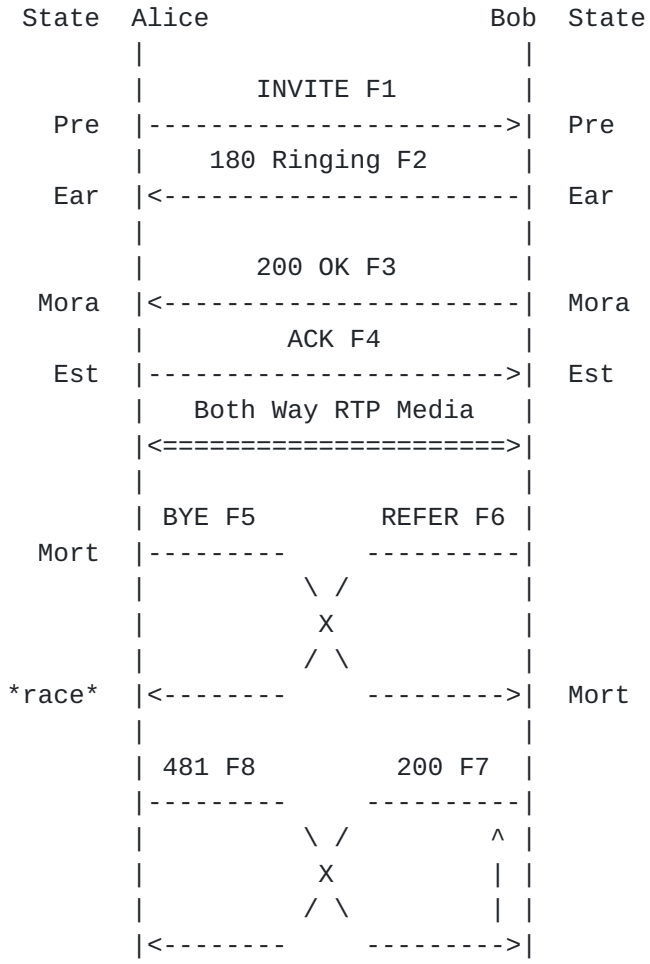
```
INVITE sip:alice@client.atlanta.example.com SIP/2.0
Via: SIP/2.0/UDP client.biloxi.example.com:5060;branch=z9hG4bKnashds7
Session-Expires: 300;refresher=uac
Supported: timer
Max-Forwards: 70
From: Bob <sip:bob@biloxi.example.com>;tag=8321234356
To: Alice <sip:alice@atlanta.example.com>;tag=9fxced76sl
Call-ID: 3848276298220188511@atlanta.example.com
CSeq: 1 INVITE
Content-Length: 0
```

```
   /* A case where a re-INVITE for a session refresh and a re-INVITE for
      a call hold are sent at the same time.  */


   F7 491 Request Pending Bob -> Alice
   /* Since a re-INVITE is in process, a 491 response are returned.  */

   F8 491 Request Pending Alice -> Bob

   F9 ACK (re-INVITE) Alice -> Bob

   F10 re-INVITE Bob -> Alice

   INVITE sip:alice@client.atlanta.example.com SIP/2.0
   Via: SIP/2.0/UDP client.biloxi.example.com:5060;branch=z9hG4bKnashds7.1
   Session-Expires: 300;refresher=uac
   Supported: timer
   Max-Forwards: 70
   From: Bob <sip:bob@biloxi.example.com>;tag=8321234356
   To: Alice <sip:alice@atlanta.example.com>;tag=9fxced76sl
   Call-ID: 3848276298220188511@atlanta.example.com
   CSeq: 2 INVITE
   Content-Type: application/sdp
   Content-Length: 133

   v=0
   o=bob 2890844527 2890844527 IN IP4 client.biloxi.example.com
   s=-
   c=IN IP4 192.0.2.201
   t=0 0
   m=audio 3456 RTP/AVP 0
   a=rtpmap:0 PCMU/8000

   /* Since Bob is not the owner of Call-ID, Bob sends an INVITE again
      after 0.0-2.0 seconds.  */


   F11 200 OK Alice -> Bob

   F12 ACK Bob -> Alice

   F13 UPDATE Alice -> Bob

   UPDATE sip:sip:bob@client.biloxi.example.com SIP/2.0
   Via: SIP/2.0/UDP client.atlanta.example.com:5060;branch=z9hG4bK74bf9.1
   Max-Forwards: 70
   From: Alice <sip:alice@atlanta.example.com>;tag=9fxced76sl
   To: Bob <sip:bob@biloxi.example.com>;tag=8321234356
   Call-ID: 3848276298220188511@atlanta.example.com
```

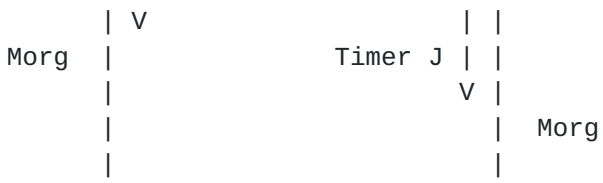
```
   CSeq: 3 UPDATE
```

```
   Content-Length: 147

   v=0
   o=alice 2890844526 2890844527 IN IP4 client.atlanta.example.com
   s=-
   c=IN IP4 192.0.2.101
   t=0 0
   m=audio 49172 RTP/AVP 0
   a=rtpmap:0 PCMU/8000
   a=sendonly

   /* Since Alice is the owner of the Call-ID, Alice sends the UPDATE
      again after 2.1-4.0 seconds.  */


   F14 200 OK Bob -> Alice
```

3.3.3 Receiving REFER (Establish state) in Mortal state



```
          | ^                    | |
          | | Timer K            | |
```



This scenario illustrates the race condition which occurs when UAS receives an Established message, REFER, in the Mortal state. Bob sends a REFER, and Alice sends a BYE at the same time. Bob send the REFER in the same dialog. Alice sends an error response to the requests which operates the session, such as REFER, because by sending the BYE, Alice had terminated the session which would have corresponded to the REFER. For handling of dialogs with multiple usages, as can be seen in the use of REFER method, see the draft on dialog usage [8].

Message Details

F1 INVITE Alice -> Bob

F2 180 Ringing Bob -> Alice

F3 200 OK Bob -> Alice

F4 ACK Alice -> Bob

F5 BYE Alice -> Bob
/* Alice sends a BYE request and terminates the session, and transits
   from the Confirmed state to Terminated state. */

F6 REFER Bob -> Alice

```
REFER sip:alice@client.atlanta.example.com SIP/2.0
Via: SIP/2.0/UDP client.biloxi.example.com:5060;branch=z9hG4bKnashds7
Max-Forwards: 70
From: Bob <sip:bob@biloxi.example.com>;tag=8321234356
To: Alice <sip:alice@atlanta.example.com>;tag=9fxced76sl
Call-ID: 3848276298220188511@atlanta.example.com
Refer-To: sip:carol@cleveland.example.org
Contact: <sip:bob@client.biloxi.example.com;transport=udp>
CSeq: 1 REFER
Content-Length: 0
```

/* Alice sends a BYE, and Bob sends a REFER at the same time.
   Bob sends the REFER on the INVITE dialog.
   The dialog state transits to the Mortal state at the moment
   Alice sends the BYE, but Bob doesn't know it until he receives

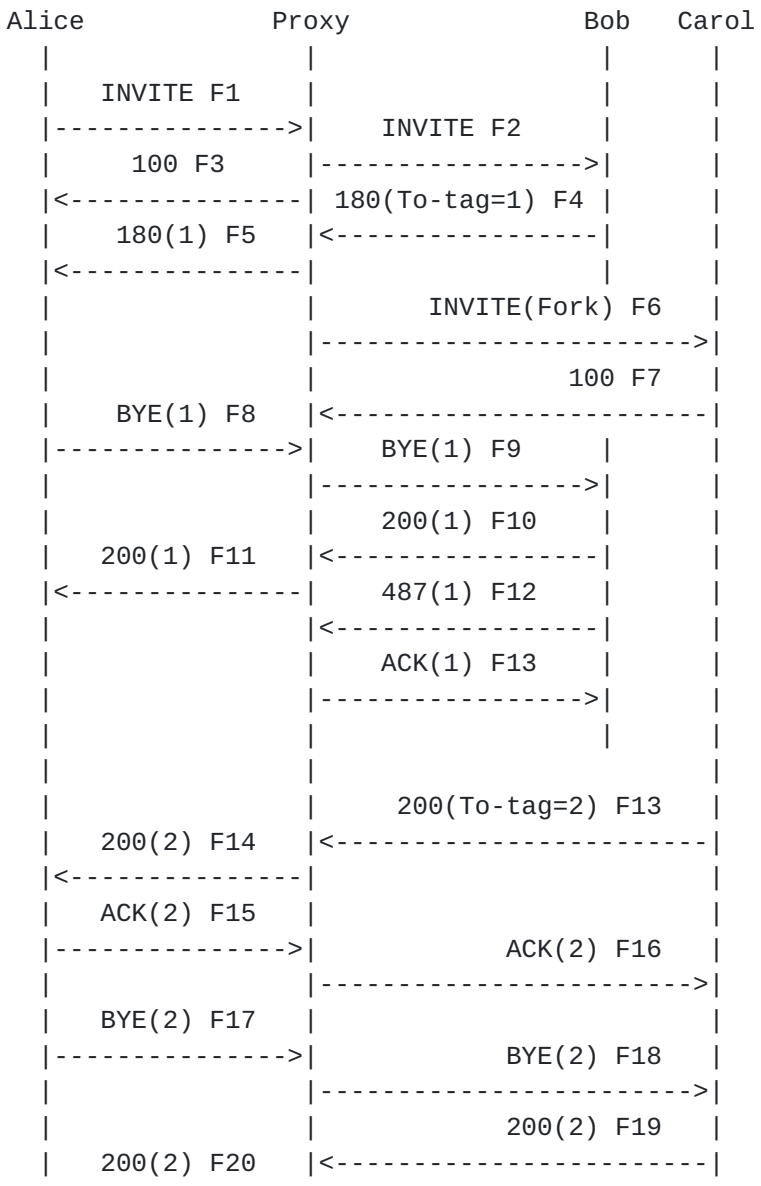
```
      the BYE.  A race condition occurs.  */
```

F7 200 OK Bob -> Alice

F8 481 Call/Transaction Does Not Exist Alice -> Bob
/* Since Alice has terminated the session, she responds with a 481
to the REFER. */

Appendix A - BYE on the Early Dialog

This section, related to Section 3.1.3, explains why BYE is not recommended in the Early state, illustrating the case in which BYE in the Early dialog triggers confusion.

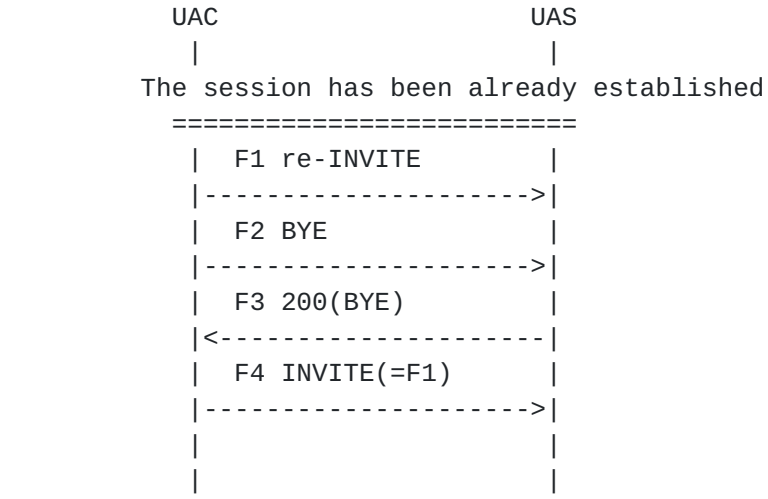


Care is advised in sending BYE in the Early state when forking by a proxy is expected. In this example, the BYE request progresses normally, and it succeeds in correctly terminating the dialog with Bob. After Bob terminates the dialog by receiving the BYE, he sends 487 to the ini-INVITE. According to Section 15.1.2 of RFC3261 [1], it is RECOMMENDED for UAS to generate 487 to any pending requests after receiving BYE. In the example, Bob sends 487 to ini-INVITE since he receives BYE while the ini-INVITE is in pending state.

However, Alice receives a final response to INVITE (a 200 from Carol) even though she has successfully terminated the dialog with Bob. This means that, regardless of the success/failure of the BYE in the Early state, Alice MUST be prepared for the establishment of a new dialog until receiving the final response for the INVITE and terminating the INVITE transaction.

The choice of BYE or CANCEL in the early state must be made carefully. CANCEL is appropriate when the goal is to abandon the call attempt entirely. BYE is appropriate when the goal is to abandon a particular early dialog while allowing the call to be completed with other destinations. When using either BYE or CANCEL the UAC must be prepared for the possibility that a call may still be established to one (or more) destinations.

Appendix B - BYE request overlapped on re-INVITE

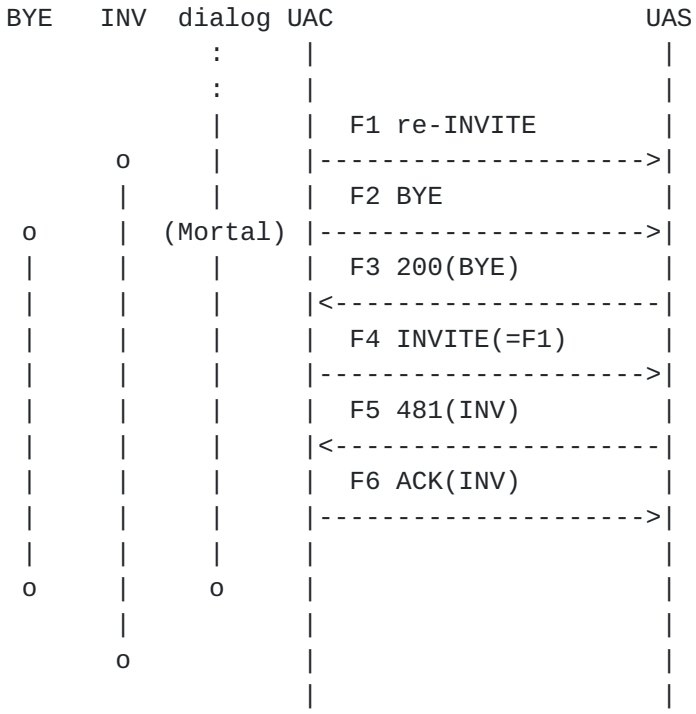


This case could look similar to the one in Section 3.2.3. However, it is not a race condition. This case describes the behavior where there is no response for INVITE for some reasons. The appendix

explains the behavior in such case and its rationale behind, since

this case is likely to cause confusion.
First of all, it is important not to confuse the behavior of the transaction layer and that of the dialog layer. RFC3261 [1] details the Transaction layer behavior. The dialog layer behavior is explained in this document.
It has to be noted that these behaviors are independent of each other, even though the both layers change their states triggered by sending or receiving of the same SIP messages (A dialog can be terminated even though a transaction still remain, and vice versa).
In the sequence above, there is no response for F1, and F2 (BYE) is sent immediately after F1 (F1 is a mid-dialog request. If F1 was ini-INVITE, BYE could not be sent before UAC received a provisional response to the request with To-tag).

Below is a figure which illustrates UAC's dialog state and transaction state.

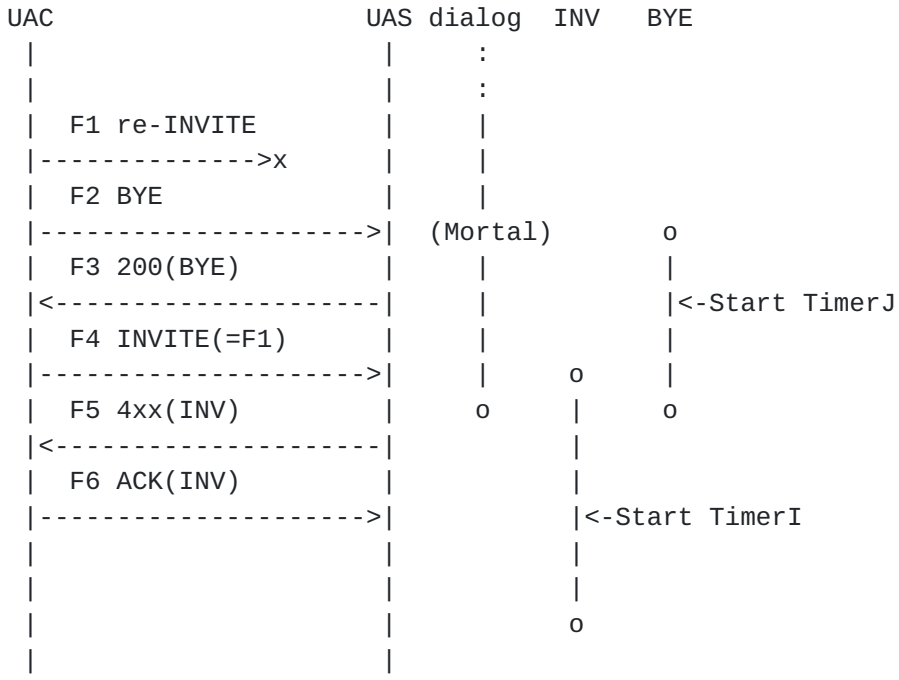


For UAC, the INVITE client transaction begins at the point F1 is sent. The UAC sends BYE (F2) immediately after F1. This is a legitimate behavior. (Usually the usage of each SIP method is independent, for BYE and others. However, it should be noted that it is prohibited to send a request with a SDP offer while the previous offer is in progress.)
After that, F2 triggers the BYE client transaction. At the same time, the dialog state transits to the Mortal state and then only a BYE or

its response can be handled.
It is permitted to send F4 (a retransmission of INVITE) in the Mortal

state, because the retransmission of F1 is handled by the transaction layer, and the INVITE transaction has not yet transited to the Terminated state. As it is mentioned above, the dialog and the transaction behave independently each other.
Therefore the transaction handling has to be continued even though the dialog moved to the Terminated state.

Next, UAS's state is shown below.



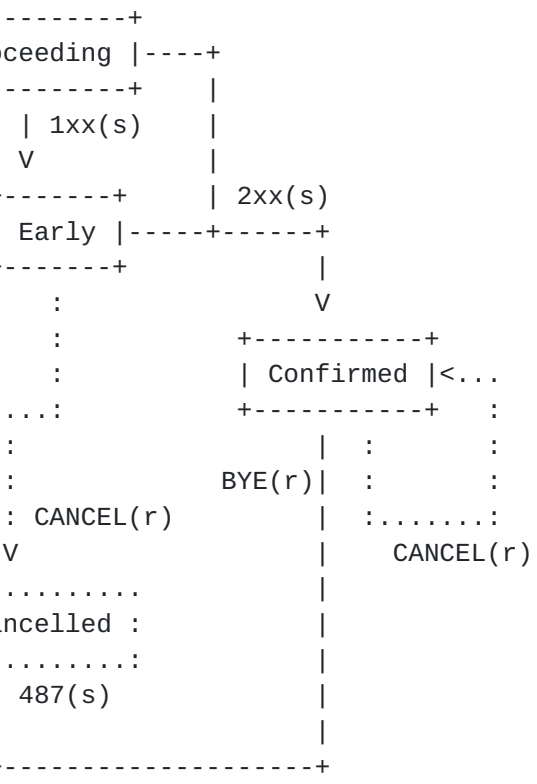
For UAS, it can be regarded that F1 packet is lost or delayed (Here the behavior is explained for the case UAS receives F2 BYE before F1 INVITE). Therefore, F2 triggers the BYE transaction for UAS, and simultaneously the dialog moves to the Mortal state.
Then, upon the reception of F4 the INVITE server transaction begins. (It is allowed to start the INVITE server transaction in the Mortal state. The INVITE server transaction begins to handle received SIP request regardless of the dialog state.)
UAS's TU sends an appropriate error response (probably 481) for F4 INVITE, because the TU knows that the dialog which matches to the INVITE is in the Terminated state.
(It is mentioned above that F4 (and F1) INVITE is a mid-dialog request. Mid-dialog requests have a To-tag. It should be noted that UAS's TU does not begin a new dialog upon the reception of INVITE with a To-tag.)

Appendix C - UA's behaviour for CANCEL

This section explains the CANCEL and the Expires header behaviors which indirectly involve in the dialog state transition in the Early state. CANCEL does not have any influence on UAC's dialog state. However, the request has indirect influence on the dialog state transition because it has a significant effect on ini-INVITE. Similarly, the Expires header does not have direct influence on the dialog state transition, but it indirectly affect the state transition because its expiration triggers the sending of CANCEL. For UAS the CANCEL request and the Expires header timeout have more direct effects on the dialog than the sending of CANCEL by UAC, because they can be a trigger to send the 487 response. Figure 3 explains UAS's behavior in the Early state. This flow diagram is only an explanatory figure, and the actual dialog state transition is as illustrated in Figure 1 and 2.

In the flow, full lines are related to dialog state transition, and dotted lines are involved with CANCEL. (r) represents the reception of signals, and (s) means sending. There is no dialog state for CANCEL, but here the Cancelled state is virtually handled just for the ease of understanding of the UA's behavior when it sends and receives CANCEL.

Below, UAS's flow is explained.



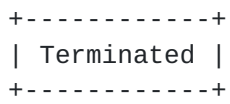


Figure 3. CANCEL flow diagram for UAS

There are two behaviors for UAS depending on the state when it receives CANCEL.
One is when UAS receives CANCEL in the Proceeding and Early states. In this case the UAS immediately sends 487 for the INVITE, and the dialog transits to the Terminated state.
The other is the case in which UAS receives CANCEL in the Confirmed state. In this case the dialog state transition does not occur because UAS has already sent a final response to the INVITE to which the CANCEL is targeted.
(Note that, from the point of UAC's behavior, it can be expected that UAS receives BYE immediately after the reception of CANCEL and moves to the Terminated state. However, the UAS's state does not transit until it actually receives BYE.)

Appendix D - Notes on the request in Mortal state

This section describes UA's behavior in the Mortal state which need careful attention.

In the Mortal state, BYE can be accepted, and the other messages in the INVITE dialog usage are responded with an error. However, sending of ACK and the authentication procedure for BYE are conducted in this state.
(The handling of messages concerning multiple dialog usages is out of the scope of this document. Refer to [8] for further information.)

ACK for error responses is handled by the transaction layer, so the handling is not related to the dialog state. Unlike the ACK for error responses, ACK for 2xx responses is a request newly generated by TU. However, the ACK for 2xx and the one for error responses are both a part of the INVITE transaction, even though their hadlings differ (Section 17.1.1.1, RFC3261 [1]).
Therefore, the INVITE transaction is completed by the three-way handshake, which includes ACK, even in the Mortal state.
Considering actual implementation, UA needs to keep the INVITE dialog usage until the Mortal state finishes, so that it is able to ACK for a 2xx response in the Mortal state.
If a 2xx to INVITE is received in the Mortal state, the duration of the INVITE dialog usage will be extended to 64*T1 seconds after the receiving of the 2xx, to cope with the possible 2xx retransmission.
(The duration of the 2xx retransmission is 64*T1, so the UA need to

be prepared to handle the retransmission for this duration.)

However, the UA shall send error response to other requests, since the INVITE dialog usage in the Mortal state is kept only for the sending of ACK for 2xx.

BYE authentication procedure shall be processed in the Mortal state. When authentication is requested by 401 or 407 response, UAC resends BYE with an appropriate credentials. Also UAS handles the retransmission of the BYE which it requested authentication itself.

References

[1] Rosenberg, J., Schulzrinne, H., Camarillo, G., Johnston, A., Peterson, J., Sparks, R., Handley, M. and E. Schooler, "SIP: Session Initiation Protocol", RFC 3261, June 2002.

[2] Rosenberg, J. and H. Schulzrinne, "An Offer/Answer Model with SDP", RFC 3264, April 2002.

[3] Johnston, A., Donovan, S., Sparks, R., Cunningham, C. and K. Summers, "Session Initiation Protocol (SIP) Basic Call Flow Examples", BCP 75, RFC 3665, December 2003.

[4] Johnston, A., Donovan, S., Sparks, R., Cunningham, C. and K. Summers, "Session Initiation Protocol (SIP) Public Switched Telephone Network (PSTN) Call Flows", BCP 76, RFC 3666, December 2003.

[5] Sparks, R., "The Session Initiation Protocol (SIP) Refer Method", RFC 3515, April 2003.

[6] Rosenberg, J. and H. Schulzrinne, "Reliability of Provisional Responses in the Session Initiation Protocol (SIP)", RFC 3262, June 2002.

[7] Rosenberg, J., Schulzrinne, H., Mahy, R., "An INVITE-Initiated Dialog Event Package for the Session Initiation Protocol (SIP)", RFC 4235, November 2005.

[8] Sparks, R., "Multiple Dialog Usages in the Session Initiation Protocol", draft-ietf-sipping-dialogusage-05 (work in progress), November 9, 2006.

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