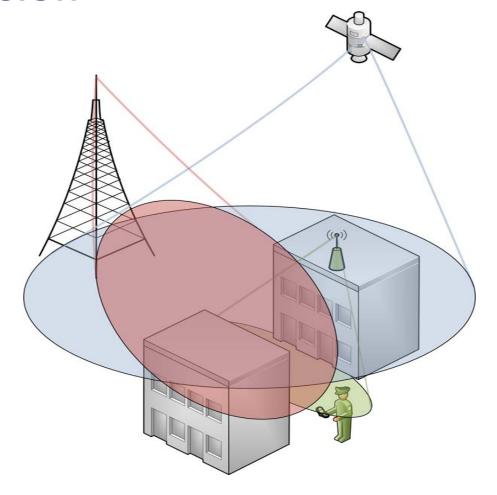
Mobile SCIP: Managing Terminal Mobility in Heterogeneous Networks with the SIP Handover Extension

Elin Sundby Boysen

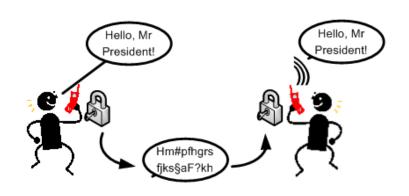
Norwegian Defence Research Establishment (FFI)

June 22, 2010

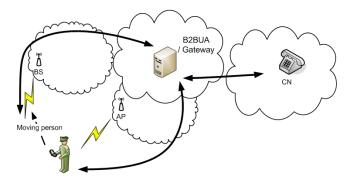




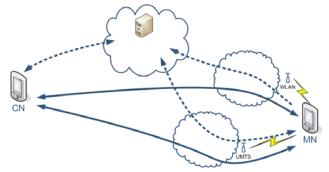
This presentation will introduce the SIP Handover Extension and its importance for SCIP.



Secure communication with SCIP

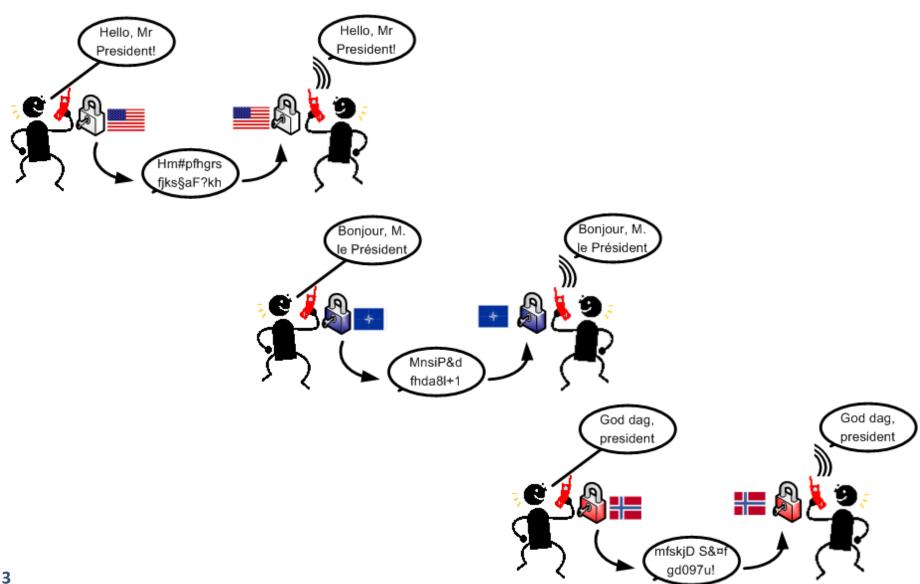


Mobile SCIP and implications of handover delay

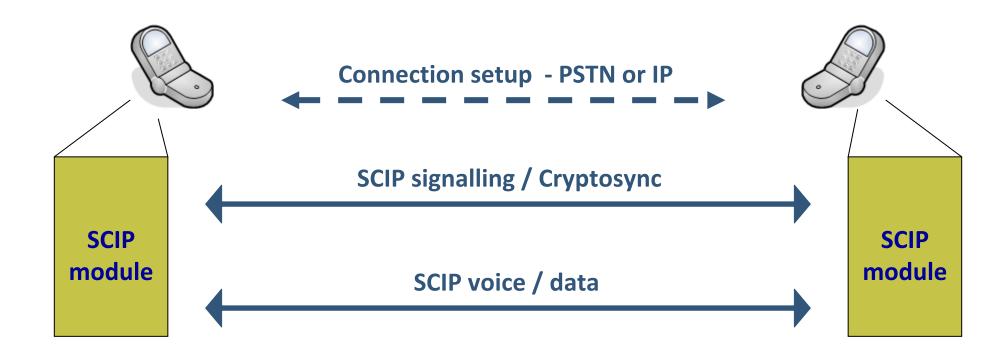


Reducing the handover delay using the SIP Handover Extension

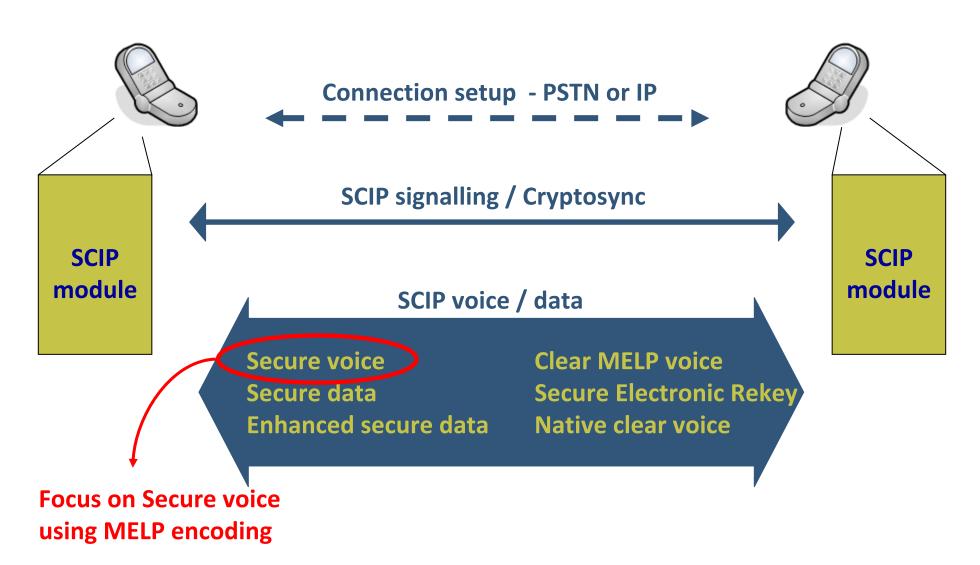
Secure Communications Interoperability Protocol (SCIP) supports both national sovereignty and interoperability.



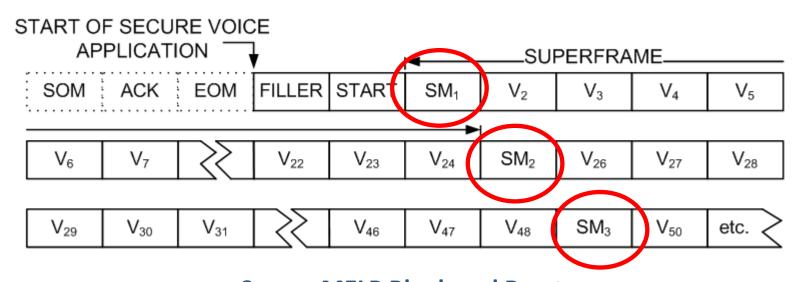
A network connection is established before control is passed to a SCIP module.



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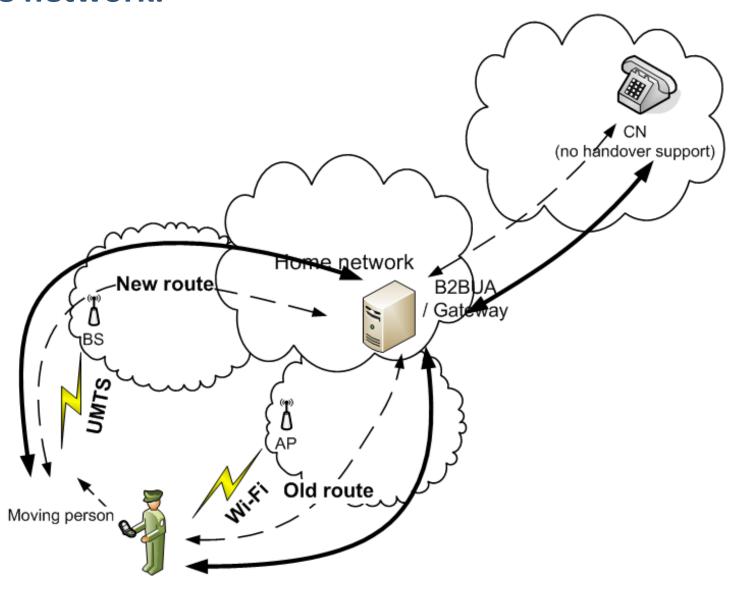
MELP is the minimum requirement for speech coding in SCIP.



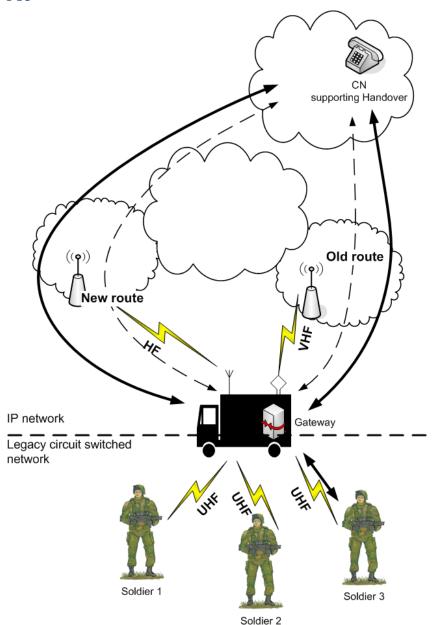
Secure MELP Blank and Burst

Minimum channel capacity requirement in SCIP is 2400 bit/s

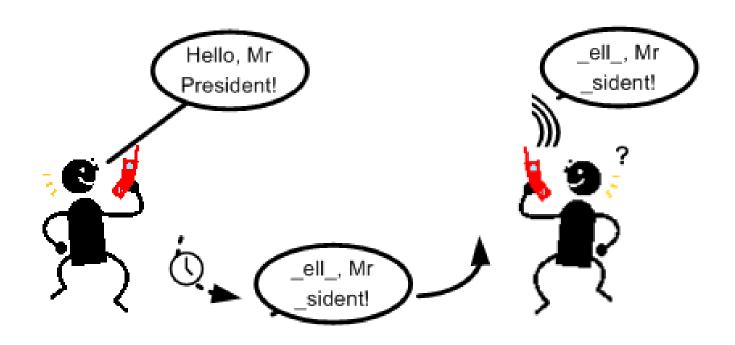
A handover can happen either in the end node or in a part of the network.



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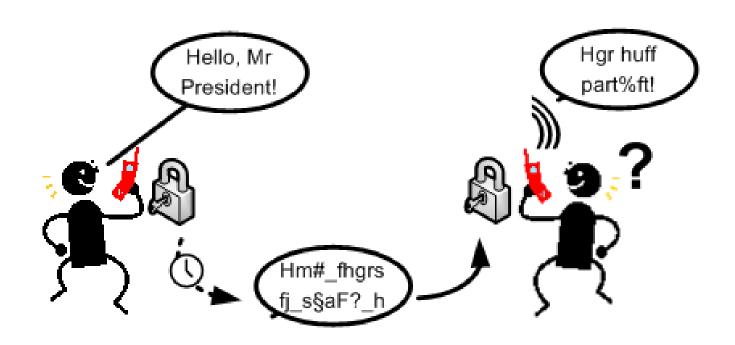
During a handover, packets can be lost leading to poor user experience.



Worst case handover delay: 450 ms [Nakajima et al. 2003]

Delay should not exceed 200 ms [ETSI 2002]

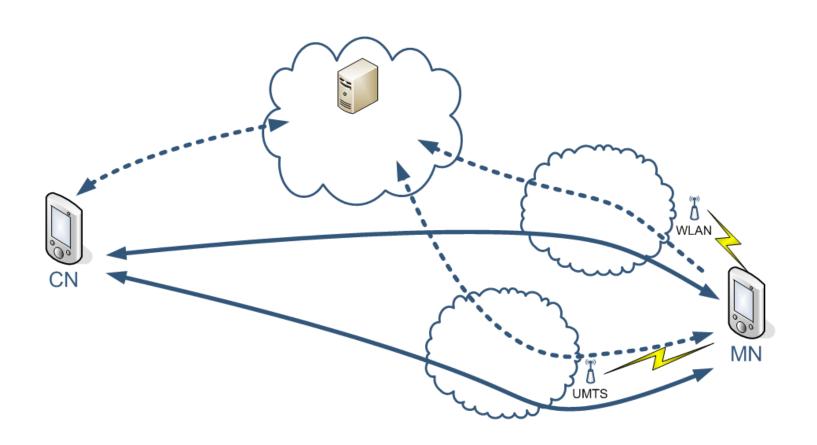
When using secure MELP, lost packets can give even worse results due to lost SYNC management frames.



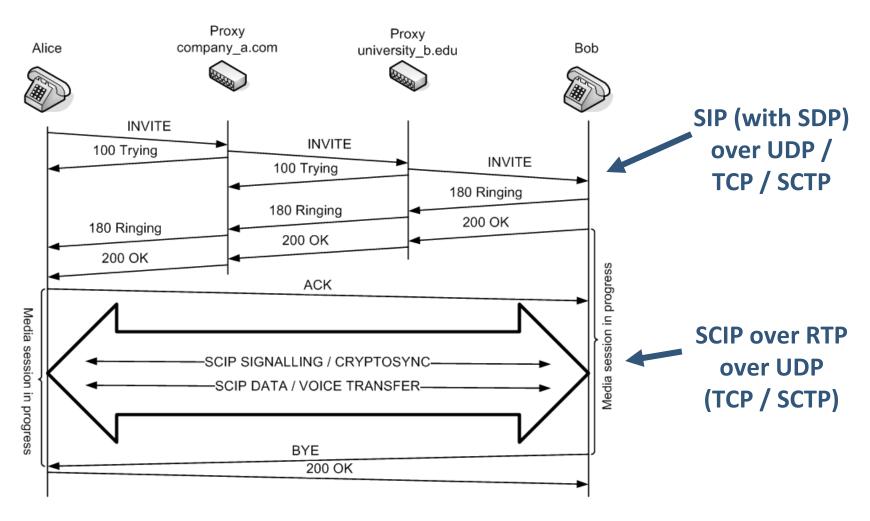
Worst case due to total resync: 1620 ms

Delay should not exceed 200 ms [ETSI 2002]

In a heterogeneous network there is usually no centralized mobility management.

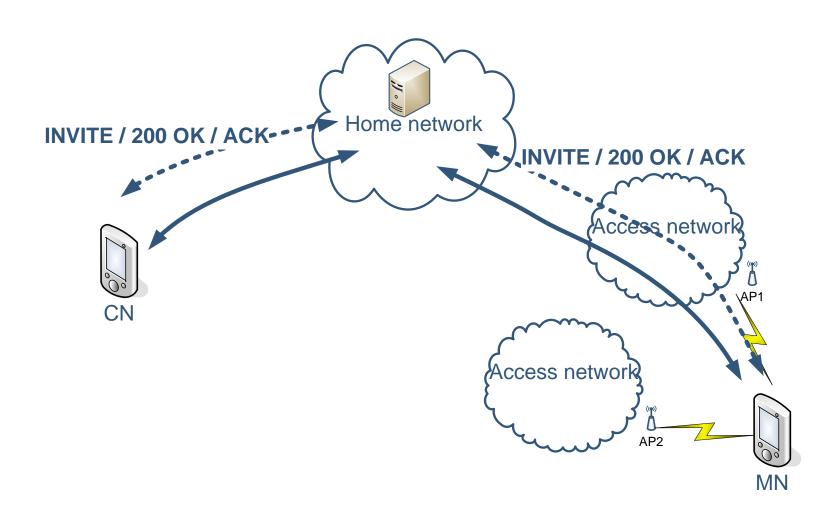


For SCIP over IP/RTP, SIP has been proposed as the signalling protocol.

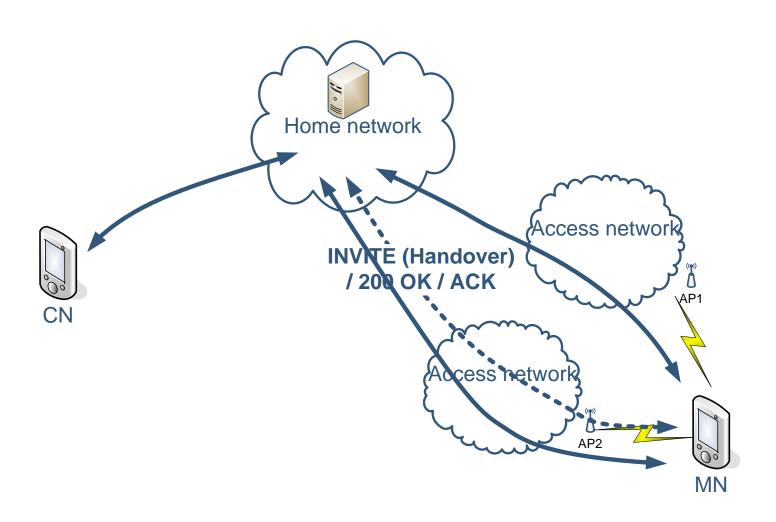


SIP = Session Initiation Protocol

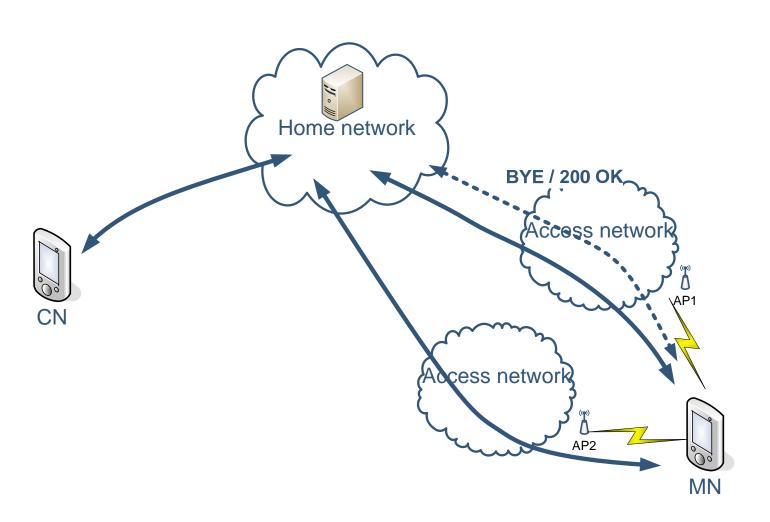
The suggested SIP extension—the Handover Extension—will eliminate packet loss during handover.



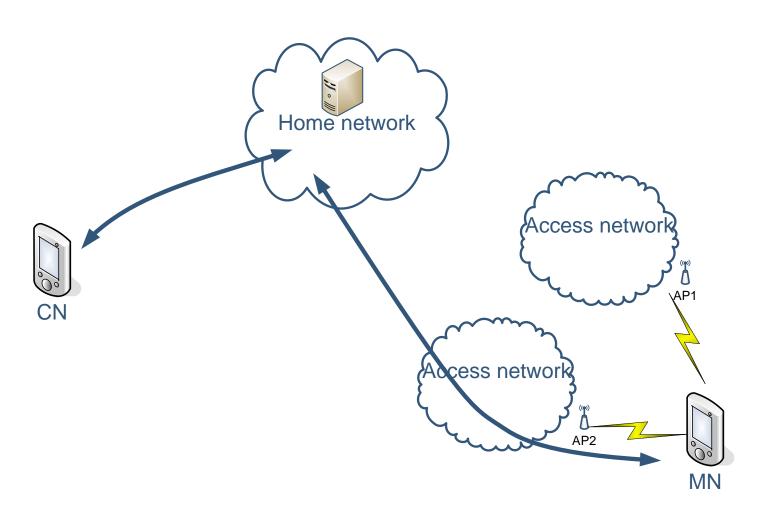
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nple message with the SIP Handover Extension.

INVITE sip:bob@biloxi.com SIP/2.0

Via: SIP/2.0/UDP

pc33.atlanta.com;branch=z9hG4bK776asdhds

Max-Forwards: 70

To: Bob <sip:bob@biloxi.com>

From: Alice <sip:alice@atlanta.com>;tag=1928301774

Call-ID: a84b4c76e66710@pc33.atlanta.com

Handover: Call-Id=33d9f110cdb0@193.156.96.196;

To-tag=5f7b910a; From-tag=as14ff55c1

CSeq: 314159 INVITE

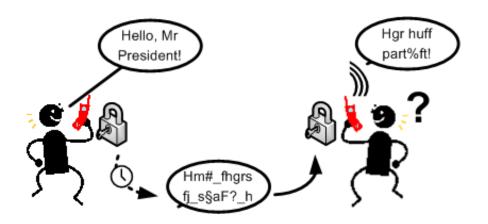
Contact: <sip:alice@pc33.atlanta.com>

Content-Type: application/sdp

Content-Length: 142

mmary, mobility for SCIP is an issue that must be essed.

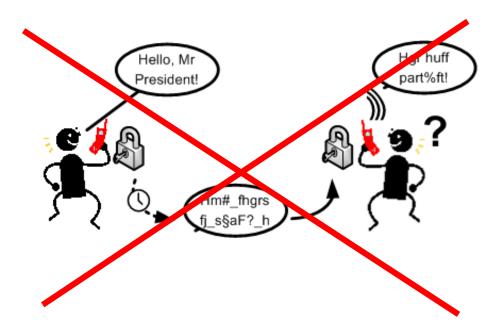
over delays can have impact in SCIP due IC management



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P Handover Extension les seamless handover terogeneous networks



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