

SIP Intrusion Detection and Prevention: Recommendations and Prototype

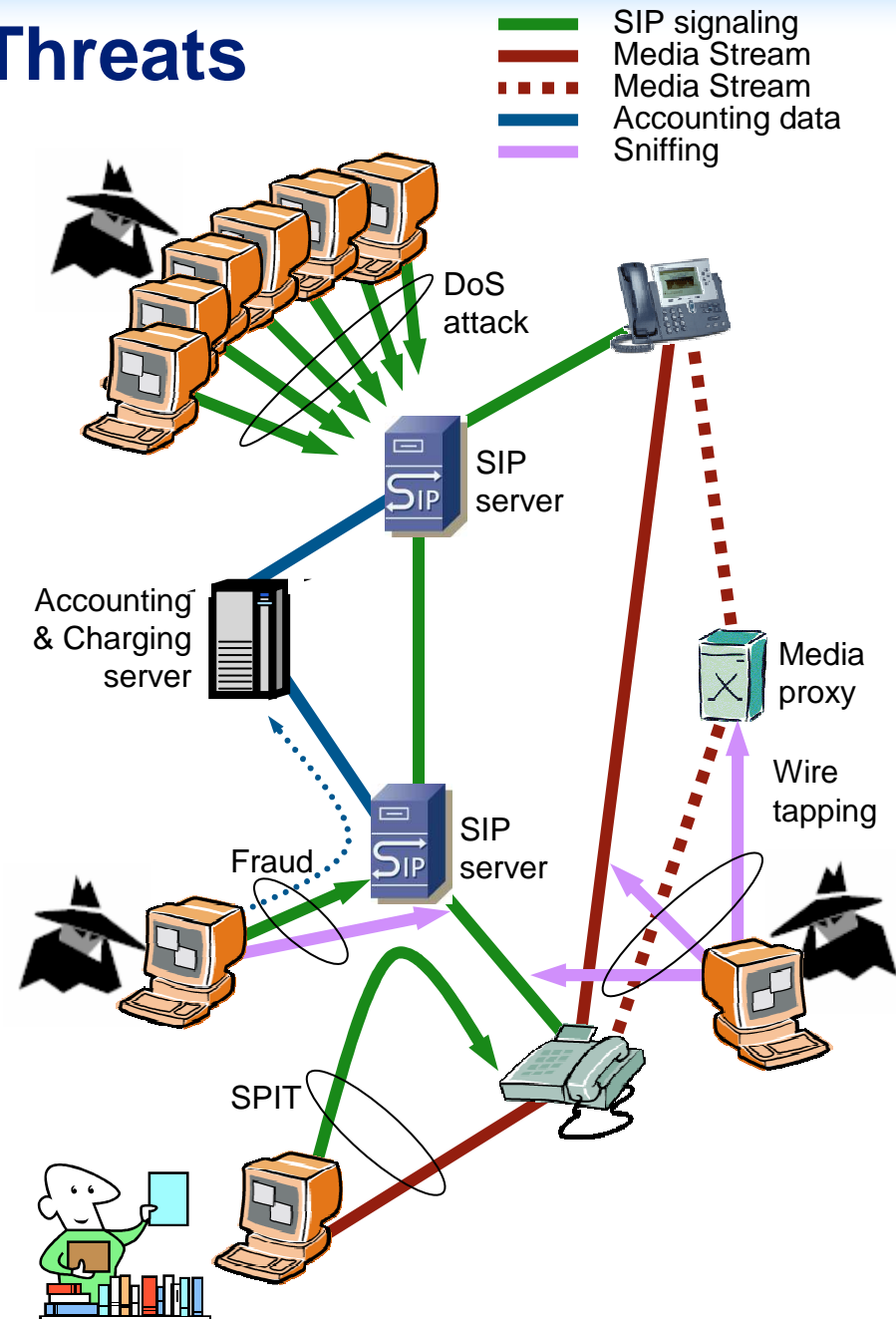
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Security Threats

- **VoIP protocols (like SIP) are vulnerable to many attacks**
 - Interruption of Service attacks (Denial of Service, DoS)
 - attacks against infrastructures and terminals
 - Social attacks (SPam over Internet Telephony, SPIT)
 - disturbances and interruptions of work by ringing phone for unsolicited calls
 - Fraud
 - placing calls on other customer's bills, etc.
 - Interception (wire tapping) and Modification of calls
 - conversations may be intercepted (lack of confidentiality)
 - conversations may be modified (lack of integrity)
- see VOIPSA taxonomy



Security Threats: available solutions?

- **Standard security mechanisms**
 - See talk of this morning by Cullen Jennings
- **Intrusion Detection and Prevention systems (IDS/IPS) needed on top of such standards**
 - to better secure the deployment
 - block the attackers bypassing security mechanisms
 - should control both signaling and media path
 - correlation needed among the two paths
 - media communication can be routed independently of the call setup path
 - three common types of IDS systems
 - host-based
 - network-based
 - stack-based
 - IPS systems can take immediate action
 - techniques
 - signatures (knowledge-based)
 - statistical observation (behavior-based)

VoIP Intrusion detection and prevention

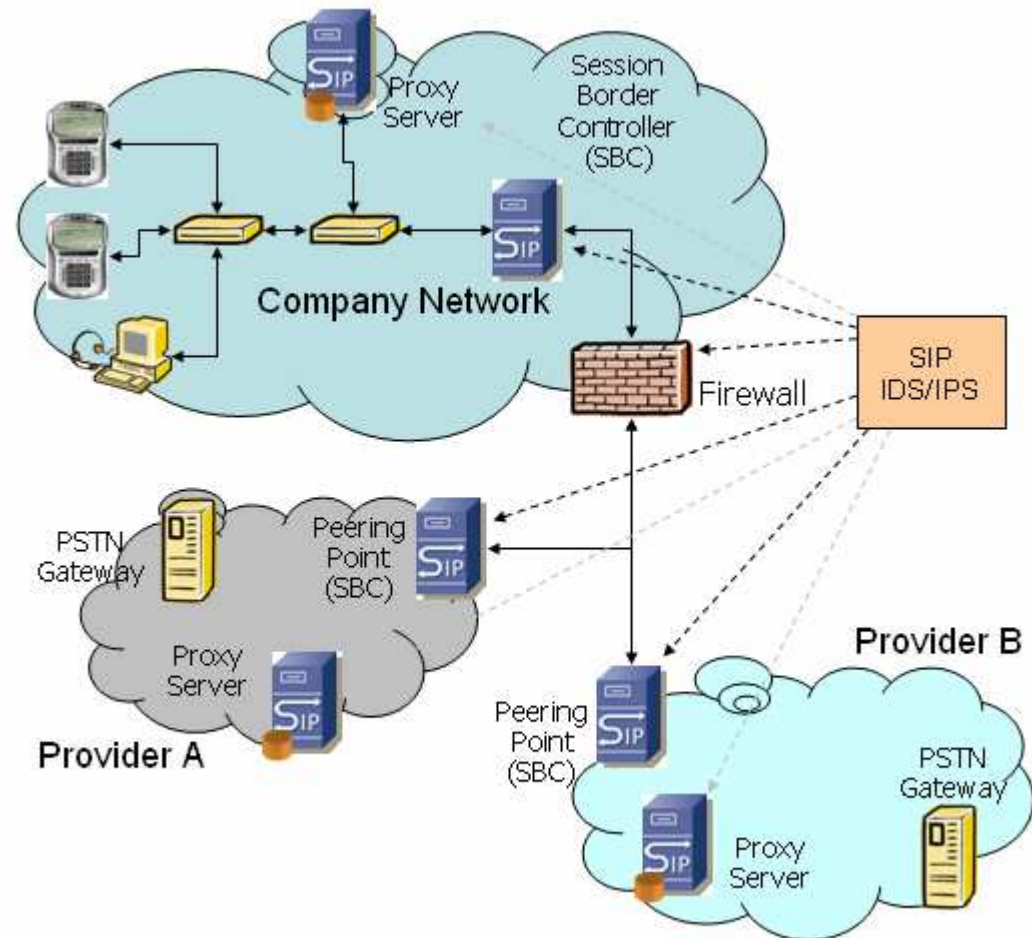
- **Attacks to traditional server oriented applications**
 - the security target is only the server
 - e.g. HTTP, FTP, E-MAIL
- **VoIP deployments have different characteristics**
 - a much higher number of systems to be protected
 - VoIP servers, e.g. Proxies and Gateways
 - terminals
 - stricter requirements in terms of security checks
 - no need of sending high rates of messages
 - few messages are able to cause crashes or reboots

Which IDS/IPS for VoIP?

- **Network-based IDS**
 - good matching of requirements
- **Host-based/Stack-based**
 - not scalable (unless you want to protect only some servers)
- **Techniques**
 - knowledge-based first
 - blocking malicious traffic
 - behavior-based second
 - statistical analysis
- **Knowledge-based techniques share info with behavior-based**
 - writing info in a shared memory area
 - IP addresses
 - SIP URIs
 - Ports
 - Message rate
 - increase in scalability (message known to be malicious are already filtered out)
 - decrease in false positives

Network-based IDS

- **Must be implemented in devices able to observe the traffic to be analyzed**
 - the entry point of the SIP network is the most suited point
- **SIP devices**
 - SIP-aware firewall
 - Peering points
 - Session Border Controllers (SBCs)
 - B2BUA in SIP
 - SIP gateways

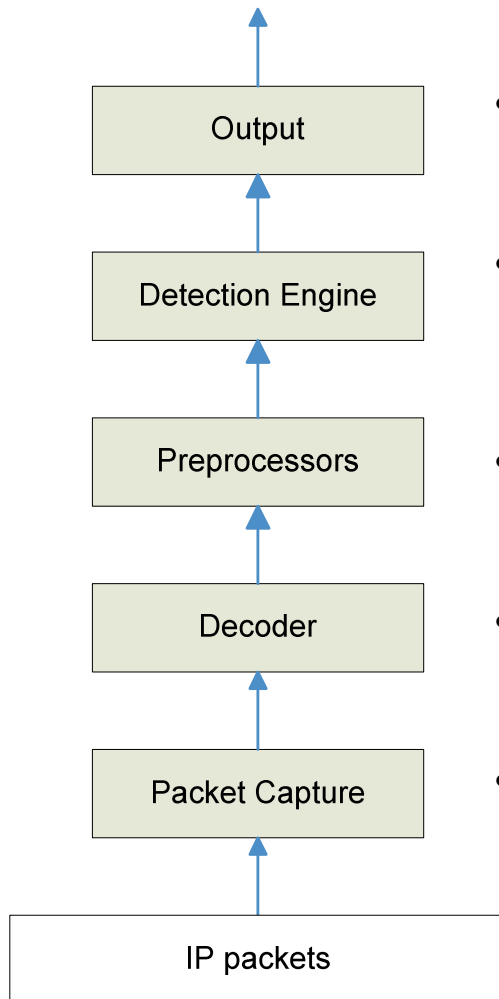


SIP IDS/IPS prototype

- We used Snort_Inline framework
- Snort is capable of performing real-time traffic analysis and packet logging
 - acts only as IDS (can only detect. not block packets)
 - it perform protocol analysis and content searching/matching
- Using Snort_Inline as IPS
 - modification and blocking of packets (accepts packets from IPTABLES using the ip_queue module)
 - works in bridge modality, invisible to attackers

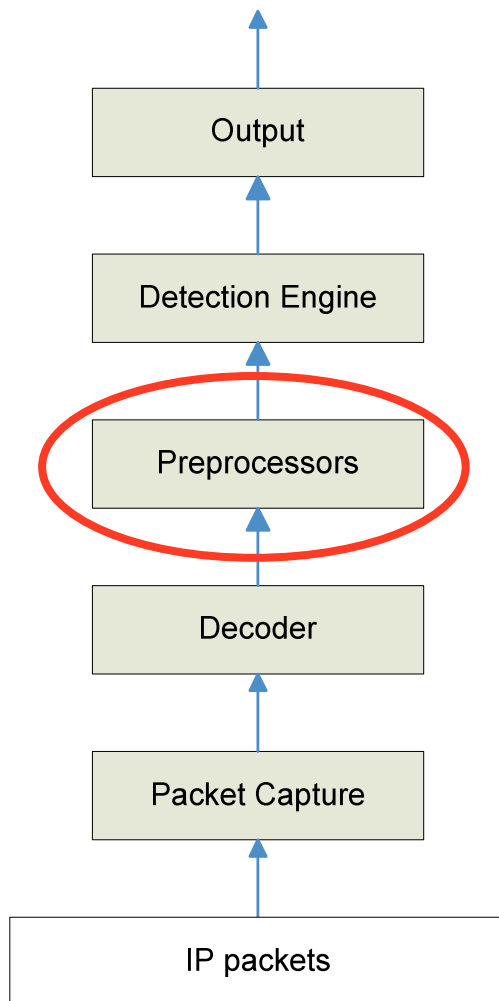


Snort architecture



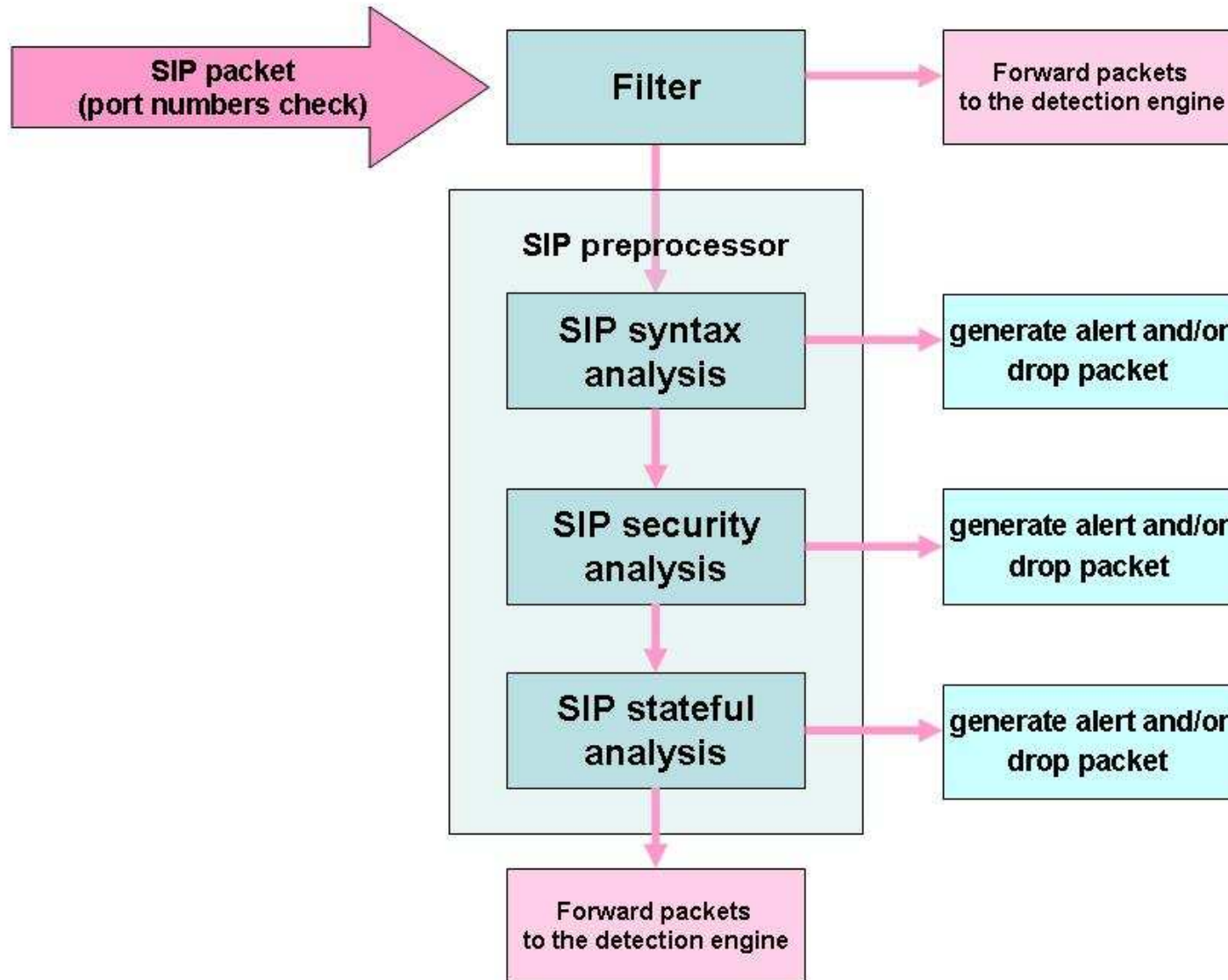
- **Output block**
 - manages the log output
 - output log is configurable (e.g. text files, databases or user-defined)
- **Detection Engine block**
 - analysis of protocols of any layer using signatures and rules
 - stateless mode
 - rule sets defined before start time
- **Preprocessors block**
 - analysis of protocols of any layer using custom made C/C++ programs
 - stateful mode
- **Decoder block**
 - syntax analysis at layer 2, 3 and 4 of the IP packet (MAC, IP and TCP/UDP)
 - Layer 2, 3 and 4 headers are inserted in a shared portion of memory
- **Packet Capture block**
 - capture the packets, it uses either libpcap or iptables depending on the Snort mode

SIP IDS/IPS prototype software (I)



- **SIP preprocessor wrote from scratch**
- **It uses oSIP libraries**
- **What it does**
 - **SIP syntax analysis (parsing)**
 - **Security check**
 - looks at mandatory fields in a SIP message
 - **Stateful analysis (soft states are used)**
 - it computes message rate and compare them to a threshold
 - by looking at SIP URIs
 - by looking at IP addresses
 - it is customizable to prevent specific DoS/SPIT attacks
 - **Generation of logs of suspicious packets in a tcpdump format**
 - can be later analyzed using Ethereal
 - can be exported to correlate with media analysis

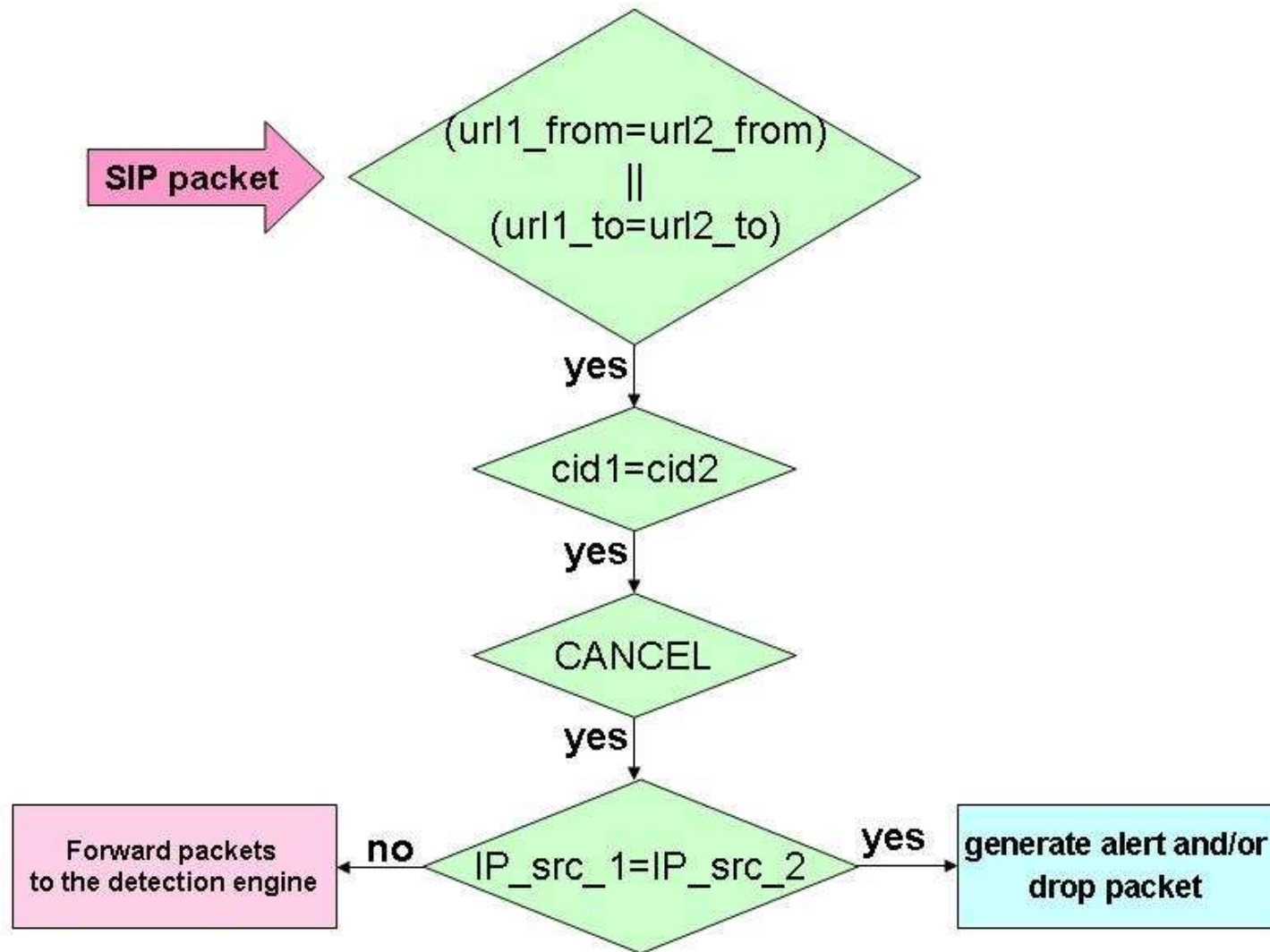
SIP IDS/IPS prototype software (II)



Examples of additional features implemented

- **Blocking SPIT attacks**
 - checking the INVITE rate
 - of a SIP URI
 - of a source IP address
 - configurable thresholds if UA or Proxy Server
- **Blocking DoS attacks**
 - checking total SIP message rate
 - of a SIP caller
 - of a source IP address
 - configurable thresholds if UA or Proxy Server
- **Blocking Call Tear-Down attacks**
 - checking that CANCEL/BYE comes from one of the parties involved in the call
 - looking at IP addresses
 - this attack can be done only spoofing To; From; Call-ID fields

Blocking Call Tear-Down attacks

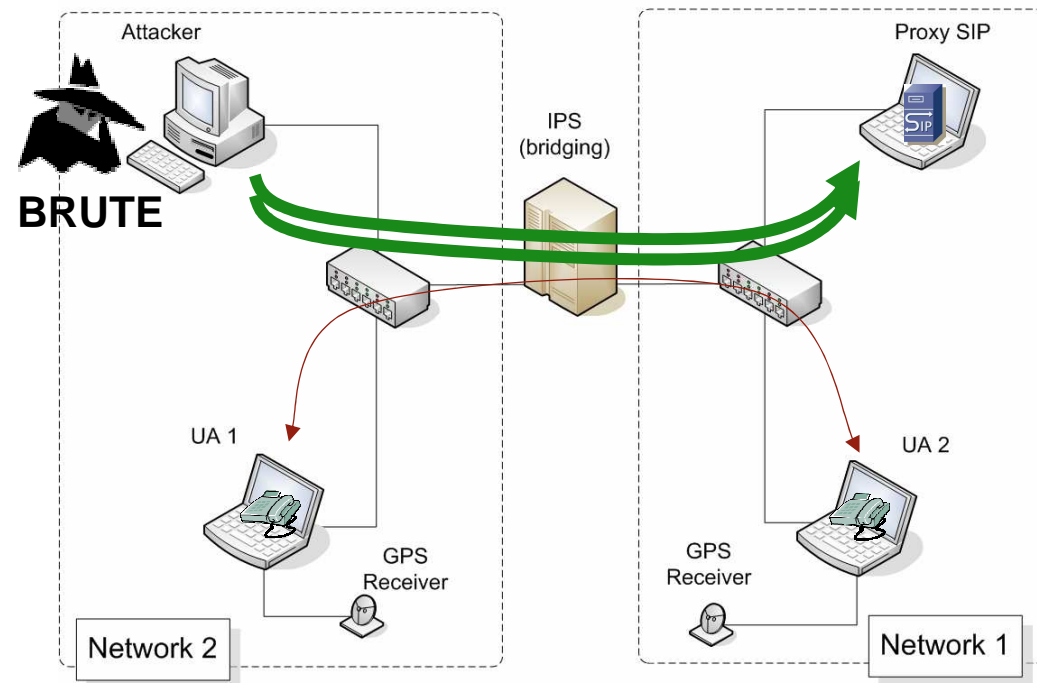


Prototype testing

- What happens to QoS of communications?
 - We stressed the IDS/IPS knowledge-based techniques generating malformed messages with different rates
 - wrote SIP plug-in generator for BRUTE
 - high performance packet generator
 - precise message rate
 - RTP media session between UA1 and UA2 at the same time
 - mean end-to-end delay
 - packet losses
 - mean jitter
 - packets with jitter > 50 ms

— SIP signaling
— Media Stream

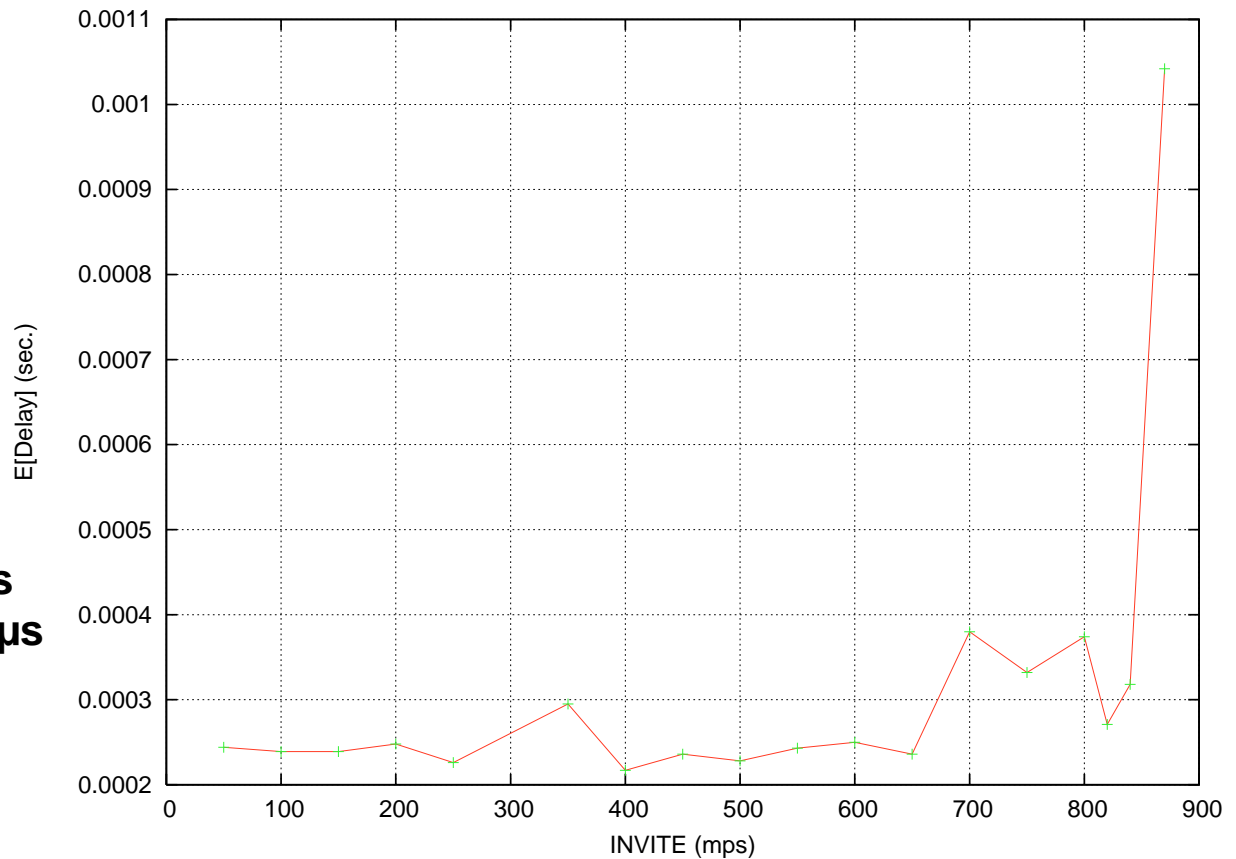
- Experimental set-up
 - SIP proxy (SIP Express Router, SER)
 - SIP UAs (Kphone)
 - running on Linux OS
 - GPS-synchronized to compute One Way Delay
 - Attacker (BRUTE generator)



End-to-end delay

- **Message generation rate lower than 860 mps**
 - mean end-to-end delay introduced by IDS/IPS ok
- **Message generation rate higher than 860 mps**
 - ip_queue module receiving packets from the iptables becomes full

- **Other parameters**
 - rate < 860 mps
 - no packet losses
 - mean jitter: 180 μ s
 - jitter > 50 ms: 10 out of 15000 packets



Conclusions/Future work

- **Guidelines for IDS/IPS for VoIP deployments**
- **Prototype implementation on top of Snort framework**
- **SIP plug-in for high performance tool (BRUTE)**
- **Evaluation of prototype implementation**

- **Future work**
 - **hybrid solution with knowledge-based checks implemented at OS kernel level (modification to iptables)**
 - **behavior-based techniques still in user space because of the flexibility required**
 - **modeling VoIP-specific DoS attacks**
 - **modeling VoIP communications**
 - **advanced stateful analysis**
 - **statistical pattern filter**
 - **signaling/media correlation**

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