

Challenges in Securing Multimedia Communications



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June 9th, 2006.



CQR



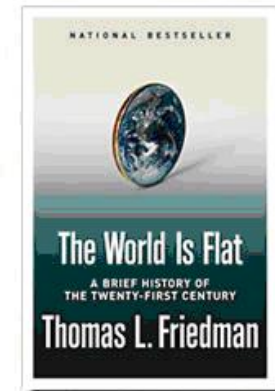
IEEE CQR 2006, London

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Why are Attacks Increasing?

Explosion of IP networking: *The world is flat*



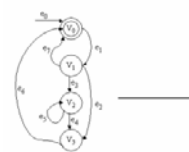
- Limited authentication/trust/encryption
- Wide variety of networked devices
- Limited security checks in software
- Limited security expertise



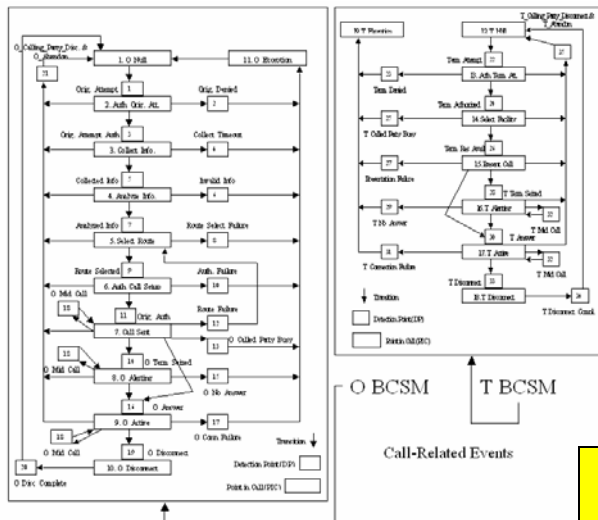
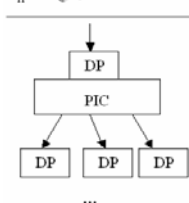
Why are Attacks Increasing?

Telecommunications, meet the Web!

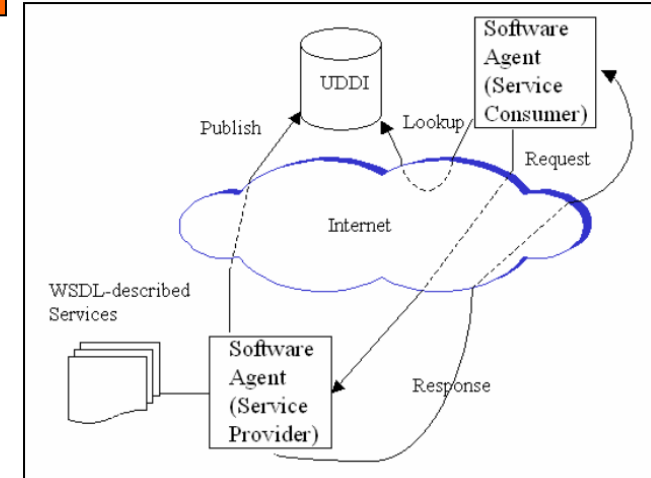
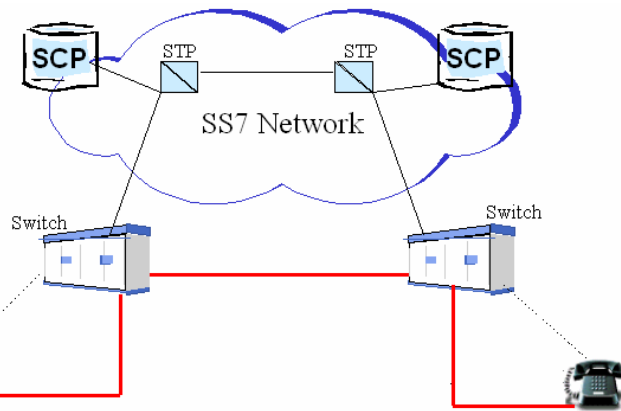
Call Models in PSTN



V_n : Vertex, or state
 e_n : Edge, or transition



Call-Related Events



The “Telecommunications SoA” will be characterized by an emphasis on Security:

- PSTN: Security by obscurity (how many crackers have an IP address? And how many have A- and F- links to PSTN gear?)
- Internet: no one knows you’re a ...
- Web services: emerging access policies, requester authentication, ...

Signaling Security and SIP

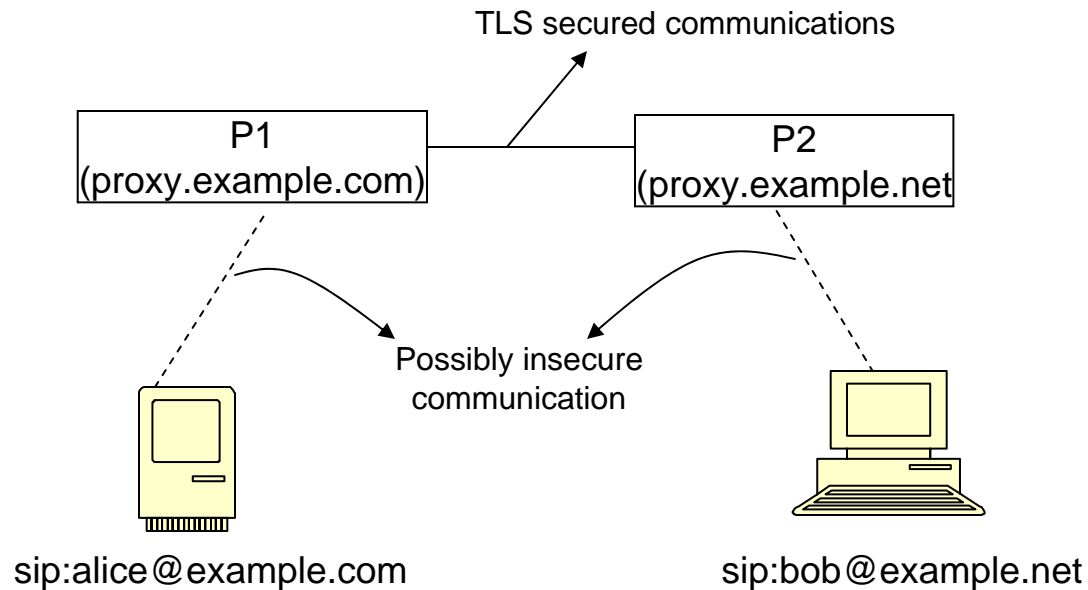
- Four ways to ensure signaling security in SIP:
 - HTTP Digest: prone to eavesdropping, replay, and MiTM attacks. Provides authentication only.
 - TLS: Hop-by-hop SIP transport security; not end-to-end! Provides confidentiality, authentication, encryption.
 - S/MIME: End-to-end signaling and body security. Provides confidentiality, authentication, encryption.
 - IPSec: Layer 3 security. Provides confidentiality and encryption.

Use of TLS in SIP references:

- [1] V.K. Gurbani and Alan Jeffrey, "The Use of Transport Layer Security (TLS) in the Session Initiation Protocol," IETF Internet-Draft, Work in Progress, February 2006, available online <<http://www.ietf.org/internet-drafts/draft-gurbani-sip-tls-use-00.txt>>
- [2] V.K. Gurbani and Alan Jeffrey, "Domain Certificates in the Session Initiation Protocol," IETF Internet-Draft, Work in Progress, February 2006, available online <<http://www.ietf.org/internet-drafts/draft-gurbani-sip-domain-certs-00.txt>>

Assumptions

Well known SIP trapezoid



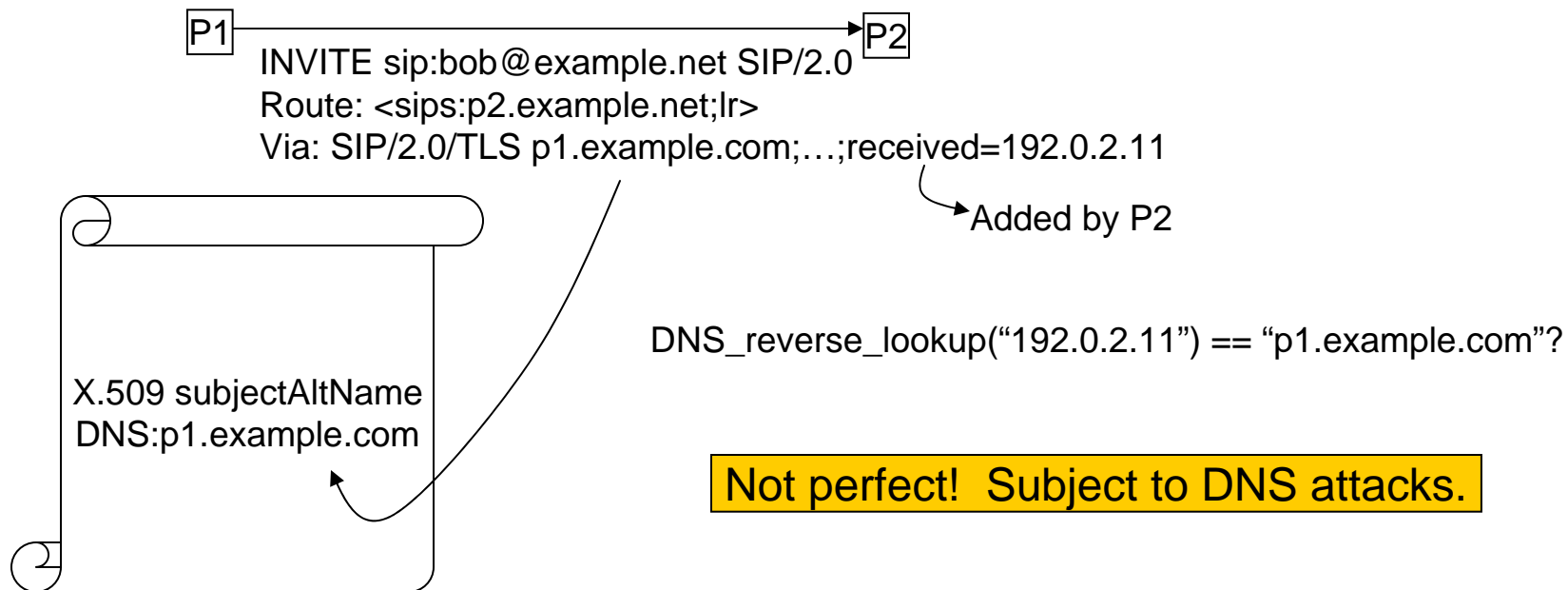
- Endpoints do not possess X.509 certificates.
- P1 and P2 support TLS and have certificates.

Open questions (#1)

- Authoritative Proxy.
 - P2 knows the request came from P1, but P2 does not know that P1 is indeed authorized to act as a proxy for the example.com domain.
 - How can this information be carried?
 - Attribute certificates (rfc3281)?
 - Trait-based authorization/SAML?
 - Existing X.509 fields?

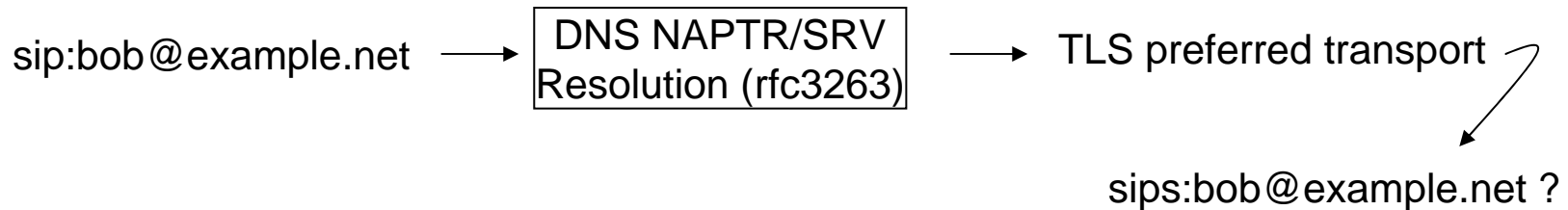
Open questions (#2)

- Mutual authentication.
 - Can rfc3261 do more on mut-auth?

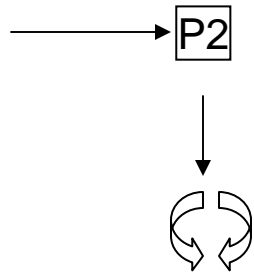


Open question (#3)

■ URI promotion.



Request arrives for
`sip:bob@example.net` but over TLS



Runs routing logic
Forward to `sip:bob@example.org`

May send over TCP

Observations:

- If Bob's paranoid, could use `sips` for forwarding.
- `example.org` domain may have configured DNS for TLS preference.

But, promotion makes the intent more explicit.

Open question (#4)

- Site certificate.

- What does it mean when multiple servers exist for a domain:

- Each server has the same high level name (example.com) in the certificate? The receiver must trust that the peer it is talking to – p1.example.com – is represented by a certificate whose DN or subjectAltName contains “example.com”.
- Each server has its canonical name (p1.example.com, p2.example.com) in the certificate?

Open question (#5)

- Leveraging the Via trail (possible use: spit)

```
INVITE sips:bob@example.net SIP/2.0
From: <sip:alice@example.org>
To: <sips:bob@example.net>
Via: SIP/TLS/2.0 egp.example.com;...
Via: SIP/TLS/2.0 proxy.aggregator.net;...
Via: SIP/TLS/2.0 uac.example.biz;...
Call-ID: 81u90—0okajyuq6
...
```

Request claims to be from example.org, but this domain does not appear in the Via trail.

Summary

- Next steps:
 - Fair amount of discussion in WG on site certificates
 - If inbound proxy presents certificate that asserts an identity of sip:example.com, then this is sufficient trust guarantee. Canonical hostname match is not required.
 - Maintain two identities in the certificate (sip:example.com and sip:p1.example.com).
 - New draft on interpreting “sips” (draft-audet-sip-sips-guidelines-00).
 - More discussion to be continued on WG mailing list.
- What does all this mean for VoIP deployments?
 - Provide a sufficient anchor of trust in the peer.
 - Lay out the rules of processing and operating assumptions to ensure minimal ambiguity during implementation.
 - Increase trust in the overall VoIP system.



Conclusion/Q&A.

Thank You!