

Monitoring SIP Traffic Using Support Vector Machines

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Abstract. We propose a novel online monitoring approach to distinguish between attacks and normal activity in SIP-based Voice over IP environments. We demonstrate the efficiency of the approach even when only limited data sets are used in learning phase. The solution builds on the monitoring of a set of 38 features in VoIP flows and uses Support Vector Machines for classification. We validate our proposal through large offline experiments performed over a mix of real world traces from a large VoIP provider and attacks locally generated on our own testbed. Results show high accuracy of detecting SPIT and flooding attacks and promising performance for an online deployment are measured.

1 Introduction

The voice over IP world is facing a large set of threats. SPAM on email systems takes a new and very annoying form on IP telephony advertising. This threat is known as SPIT (Spam over Internet Telephony). However, SPIT is not the only threat vector. The numerous software flaws in IP phones and servers affect their reliability and open the door to remotely attack previously unseen in the “stable” world of telecommunication operators (PSTN), which was based on mutual trust among few peers. Leveraging the IP to support voice communications exposes this service (voice) to the known denial of service attacks that can be easily implemented by service or network request flooding on the Internet. Resource exhaustion thus automatically finds its place against SIP proxies and back-to-back user agents, which are essential to support this critical infrastructure. The list of potential threats is huge and ranges from VoIP bots (that could spread by malware and perform distributed attacks, perform SPIT or toll fraud), to eavesdropping and Vishing (similar attack to the Phishing are using VoIP as the transport vehicle) [1].

Securing VoIP infrastructures constitutes one of the major challenges for both the operational and research communities because security by design was not a key component in the early phases of both VoIP research and development. VoIP-specific security solutions are currently required by the market because the research and standardization efforts are still trying hard to address the issues of securing and monitoring VoIP infrastructures.

Our work fits into these efforts and addresses a new monitoring approach for VoIP specific environments. Our monitoring scheme is based on Support Vector Machines for efficient classification. We continuously monitor a set of 38 features in signaling time slices and use these features as the raw input to the classification engine. A threshold based alarm generator is placed on top of the classification engine. We show that the system is both efficient and accurate and study the impact of the various features on the efficiency.

We start the presentation with a short survey on VoIP security with focus on flooding attacks and SPIT. We then give a functional description of our monitoring solution together with the definition of the 38 features computed in our system for classification (section 3). In section 4, we provide a short mathematical background of the SVM learning machine model used in the monitoring process. Offline traces inspection is presented in section 5 where we also describe the data set. Section 6 demonstrates the performances of our approach to detect different types of attacks. Related work is addressed in section 7. Section 8 concludes the paper and enumerates some future work.

2 The Threat Model

2.1 Flooding Attacks

Denial of service attacks can target the signaling plane elements (e.g. proxy, gateway, etc.) with the objective to take them down and produce havoc in the VoIP network. Such attacks are launched by either flooding the signaling plane with a large quantity of messages, malformed messages or executing exploits against device specific vulnerabilities.

The authors of [2] categorize some of these attacks based on the request URI and perform a comparative study of these ones against popular open source VoIP equipment. We adopt the same categorization, i.e.:

- UDP flooding: Since the vast majority of SIP systems use UDP as the transport protocol, a large amount of random UDP packets are sent in an attempt to congest the network bandwidth. Such attacks produce a high packet loss. Legitimate call signaling has thus a reduced probability to reach the target and to be processed.
- INVITE flooding with a valid SIP URI: The attacker calls one user/phone registered at a server/proxy. The proxy relays the calls to the phone. If the proxy is stateful it will manage a state machine for every transaction. The phone is quickly overloaded by the high rate of calls and is no more able to terminate the calls. As a result, the server is allocating resources for a long time and it will run out of memory.
- INVITE flooding with a non existent SIP URI: If the attacker doesn't know a valid SIP URI registered on the target, it can send calls to an invalid address. The proxy/server responds with an error response like "user not found". When the attack rate is higher than the server capabilities, the resources are exhausted. This type of flooding is less disturbing than the