**A****TIS-0x0000x**

ATIS Standard on

**Signature-Based Handling of Asserted Information Using Tokens (SHAKEN):**

 **SHAKEN Support of "div" PASSporT Token**

**Alliance for Telecommunications Industry Solutions**

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**Abstract**

The base SHAKEN specification provides replay-detection mechanisms to identify cases where a malicious entity attempts to masquerade as another user by replaying parts of a legitimate INVITE request. However, these mechanisms don’t cover cases where the INVITE is replayed within the short Date freshness window. This technical report describes how the mechanisms defined by draft-ietf-stir-passport-divert can be integrated within the SHAKEN framework to close this replay attack window.

**Foreword**

The Alliance for Telecommunications Industry Solutions (ATIS) serves the public through improved understanding between carriers, customers, and manufacturers. The [**COMMITTEE NAME**] Committee [**INSERT MISSION**]. [**INSERT SCOPE**].

The mandatory requirements are designated by the word *shall* and recommendations by the word *should*. Where both a mandatory requirement and a recommendation are specified for the same criterion, the recommendation represents a goal currently identifiable as having distinct compatibility or performance advantages. The word *may* denotes a optional capability that could augment the standard. The standard is fully functional without the incorporation of this optional capability.

Suggestions for improvement of this document are welcome. They should be sent to the Alliance for Telecommunications Industry Solutions, [**COMMITTEE NAME**], 1200 G Street NW, Suite 500, Washington, DC 20005.

At the time of consensus on this document, [**COMMITTEE NAME**], which was responsible for its development, had the following leadership:

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The **[SUBCOMMITTEE NAME]** Subcommittee was responsible for the development of this document.

**Revision History**

| **Date** | **Version** | **Description** | **Author** |
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# Scope & Purpose

## Scope

This document extends SHAKEN to support the PASSporT "div" extension defined in draft-ietf-stir-passport-divert.

## Purpose

The base STIR/SHAKEN replay-attack detection mechanisms are unable to distinguish between a legitimate call that is diverted by a feature such as call-forwarding, and a malicious call where the attacker attempts to masquerade as another user by replaying a legitimate To, From and Identity header within the Date freshness window. This document describes how draft-ietf-stir-passport-divert can be used to close this replay attack window.

### Document Organization

Section 4 provides an informative overview of the replay attack window that exists within the base SHAKEN framework, and describes how the PASSporT "div" extensions can be used to close the window.

Section 5 specifies the normative requirements to add support draft draft-ietf-stir-passport-divert to SHAKEN.

# Normative References

The following standards contain provisions which, through reference in this text, constitute provisions of this Standard. At the time of publication, the editions indicated were valid. All standards are subject to revision, and parties to agreements based on this Standard are encouraged to investigate the possibility of applying the most recent editions of the standards indicated below.

ATIS-0x0000x, *Technical Report*.

ATIS-0x0000x.201x, *American National Standard*.

# Definitions, Acronyms, & Abbreviations

For a list of common communications terms and definitions, please visit the *ATIS Telecom Glossary*, which is located at < <http://www.atis.org/glossary> >.

## Definitions

**AAA**: xxxx.

**Bbbb**: xxxx.

## Acronyms & Abbreviations

|  |  |
| --- | --- |
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# Overview

The baseline SHAKEN framework and the core STIR protocols from which SHAKEN is based support the end-to-end call authentication for the common 2-way call scenario where user-a calls user-b. For this case, the originating network generates a PASSporT token containing "orig", "dest" and "iat" claims to assert that the calling telephone number (TN) is authorized to be used as the originating identity for that specific call. The terminating network can then verify that the PASSporT signature is valid, and that the "dest" claim matches the target called TN, to determine with a high degree of certainty that the calling TN identifies the calling user.

However, for call scenarios where a call is diverted, the verification process becomes less certain due to the fact that the PASSporT "dest" claim no longer matches the target called TN. Based only on the SHAKEN Identity header from the first leg of the call, the verification service is unable to validate the diverted legs of the call and the associated changed telephone destinations. This document presents the solution for extending the SHAKEN framework to support these call diversion scenarios.

draft-ietf-stir-passport-div defines a PASSporT extension, "div", as a basis for accommodating the diversion or diversions that may occur for various SIP applications. The "div" extension enables a forwarding network to dynamically authenticate the forwarding TN that is diverting the call to a new "dest" TN as a call progresses. A terminating verification service (STI-VS) can then use this additional information to verify the associated TNs at each diversion of a call between the final destination TN and initial original destination TN.

What follows in this document is the specification of how the PASSporT "div" extension must be used as part of the SHAKEN framework for providing end-to-end SHAKEN validation for diverted calls..

# Normative Requirements

This section contains the normative requirements that must be supported by a SHAKEN Service Provider, STI-AS and STI-VS in order to enable end-to-end SHAKEN authentication for diverted calls.

## STI-AS Base SHAKEN Authentication

During base SHAKEN authentication, the STI-AS shall perform the authentication procedure defined in [shaken] with the exception that it populates the "shaken" PASSporT "dest" claim with the canonicalized value of the Request-URI TN.

## STI-AS Base SHAKEN Verification

On receiving an INVITE request containing a "shaken" Identity header and no "div" Identity headers, the STI-VS shall perform the verification procedures defined in [shaken], with the exception that it use the canonical value of the Request-URI TN as the locally created "dest" claim input to the PASSporT signature verification algorithm.

## STI-AS "div" Authentication

The STI-AS shall provide "div" authentication services as defined in [draft-ietf-stir-passport-divert], with the following restrictions:

* The requirement that the INVITE request must contain at least one Identity header is modified here to mandate that the INVITE contains at least one "shaken" Identity header,
* The "orig", "dest" and "div" claims shall be of type "tn",
* The "orig" claim value shall be copied from the “shaken” PASSporT "orig" claim,
* The "opt" claim shall not be used (no nesting).

Note that per draft-ietf-stir-passport-div, the "div" authentication service is not required to check for an unbroken chain of authority from the "shaken" PASSporT "dest" TN to the diverting TN before authenticating the current diversion event. As long as the "div" authentication service is authoritative for the diverting TN, then it simply adds a "div" PASSporT token containing the following claims:

* "orig" claim matches the "shaken" PASSporT "orig" claim,
* "div" claim matches the diverting TN,
* "dest" claim matches the Request-URI TN.

Any breakage in the chain of authority between "shaken" PASSporT "dest" claim and Request-URI TN will be detected by the remote verification service in the divert-to terminating network.

## STI-AS "div" Verification

On receiving an INVITE request containing a “shaken” Identity header and one or more “div” Identity headers, an STI-VS shall perform the “div” verification procedures defined in [draft-ietf-stir-passport-divert], with the following restrictions:

* The "div" PASSporT "orig", "dest" and "div" claims must be of type “tn”,
* The "div" PASSporT "opt" claim must be absent.

## In-network Call Diversion

The STI-AS shall perform "div" authentication as specified in section 5.3 for in-network call diversion (e.g., where an in-network call feature or routing function updates the canonical value of the TN contained in the Request-URI of an INVITE request).

If an originating network function updates the INVITE Request-URI TN, and the originating SP has authority over the "orig" claim in the "shaken" Identity header contained in the same INVITE request, then the STI-AS shall not perform "div" authentication. Instead, the STI-AS shall perform base SHAKEN authentication, and replace the existing "shaken" Identity header with a new "shaken" Identity header that reflects the new destination.

## End-user Device Call Diversion

Certain types of end-user devices such as SIP-PBXs are capable of diverting incoming calls (aka DID calls) back out to a new divert-to destination in the global network. The end-user device diverts the call either by responding to the incoming INVITE request with a 302 Moved Temporarily response, or by sending a new INVITE request to establish the divert-to call leg. The requirements in this section apply to the case where device capabilities and service provider policies enable the end-user device to divert calls using these mechanisms.

### Call Diversion via 302-Response

The STI-AS shall perform "div" authentication on behalf of an end-user device that diverts a call via a 302 Moved Temporarily response. In this case, the PASSporT "div” claim shall be populated with the canonicalized value of the TN contained in the 302 response Contact header.

### Call Diversion via new-INVITE Request

The STI-AS provides authentication services for INVITE requests received from an end-user device. The type of authentication performed will depend on the type of INVITE; e.g., where an originating INVITE receives SHAKEN authentication, while a diverting INVITE containing a SHAKEN Identity header receives "div" authentication. If possible, the STI-AS differentiates between these two INVITE types in order to perform the proper authentication procedure.

The capabilities of the end-user device will determine whether the STI-AS is able to differentiate between INVITE-types, and if differentiation is possible, how the STI-AS identifies the specific INVITE type. There are three different cases:

1. **Determine INVITE-type based on presence/absence of Identity header(s)**

In this case, the end-user device is capable of copying the "shaken" and "div" Identity headers received in an incoming DID INVITE request to a subsequent divert-to INVITE request. Therefore, if the host SP includes a SHAKEN Identity header in every DID INVITE requests sent to the end-user device, it can assume that any received INVITE request that contains a SHAKEN Identity header is establishing the divert-to leg of a diverted DID call.

1. **Determine INVITE-type based on other SIP header information**

In this case, the end-user device is not capable of conveying received Identity header(s) in divert-to INVITE requests. However, the end-user device supports a profile of SIP that enables the STI-AS to unambiguously distinguish between an originating and diverting INVITE request (e.g., the end-user device adds a History-Info header field entry to record the diverting event).

1. **Unable to determine INVITE type**

If an end-user device does not support either of the above mechanisms, then the STI-AS will be unable to distinguish between an originating and diverting INVITE request. In this case, the STI-AS has no choice but to perform base SHAKEN authentication for all INVITE requests received from the end-user device.

The normative requirements to support these three cases are described in the following subsections.

#### Determine INVITE-type based on presence of Identity Header(s)

This section applies to the case where the end-user device is known to convey "shaken" and "div" Identity headers via divert-to INVITE requests.

The host SP shall include at least one Identity header in all DID INVITE requests sent to the end-user device. The host SP shall populate the Identity header(s) in the DID INVITE request based on the contents of the incoming INVITE request received by the host SP from the remote originating network, as follows:

* If the received INVITE request contains one or more Identity headers, and the calling user’s identity is not private (INVITE contains no Privacy header specifying a privacy-type of "id"), then include all received Identity headers in the DID INVITE request sent to the end-user device.
* If the received INVITE request contains one or more Identity headers, but the calling user’s identity is private (INVITE contains Privacy:id), then convert all received Identity header full-form PASSporT tokens to the equivalent compact form before including in the DID INVITE request sent to the end-user device.
* If the received INVITE request does not contain an Identity header, then invoke the STI-AS to perform SHAKEN authentication on the calling TN of the INVITE request received from the remote originating network. Since the calling TN is not assigned to the host SP in this case, it will be authenticated with Gateway level attestation. If the calling user identity is private, then the compact form of the shaken PASSporT token will be sent to the end-user device in the Identity header of the DID INVITE; otherwise, the full form will be sent.

On receiving an INVITE request from an end-user device, the STI-AS shall perform "div" authentication as specified in section 5.3 if the following conditions exist:

1. The INVITE request contains a "shaken" Identity header containing a full form PASSporT token, and zero or more "div" Identity headers, and
2. The end-user device is authorized to use the diverting TN, based on the SHAKEN authentication criteria for applying full attestation defined in [shaken].

How the STI-AS identifies the diverting TN will depend on the SIP profile supported by the end-user device and the host SP network. For example, for IMS-compliant networks, the diverting TN is asserted by the P-CSCF in the P-Asserted-Identity header. In this case, before sending the INVITE request to the next hop, the STI-AS shall update the P-Asserted-Identity header to match the "orig" claim in the "shaken" Identity header.

If either of the above conditions do not exist, then the STI-AS shall perform SHAKEN authentication procedures defined in section 5.1.

*Open Issue:*

1. *Describe STI-AS procedure for handing INVITE containing compact-form SHAKEN PASSporT token. The presence of an Identity header indicates this is a diverting INVITE, but since PASSporT is compact form, STI-AS can’t identify calling TN. Therefore, it cannot create a “div” PASSporT token. One option is to add a SHAKEN full-form PASSporT token with Gateway attestation (“Gateway”, because STI-AS doesn’t know identity of the calling user). Since call is private, Verstat will not be delivered to called user, but this PASSporT token would record the event and enable subsequent trace-back.*

#### End-user Device does not Convey Identity Headers in Divert-to INVITE

This section applies to the case where the end-user device does not support the capability to convey Identity headers for diverted calls, which means that the STI-AS cannot use the presence/absence of Identity headers to distinguish between originating and diverting INVITE requests received from the end-user device. However, in this case, the end-user device supports a profile of SIP that enables the STI-AS to distinguish between originating and diverting INVITE requests based on other SIP header information.

The STI-AS shall perform both "shaken" and "div" authentication on behalf of an end-user device that diverts a call via a new INVITE request to establish the diverting call leg, if the following conditions exist:

1. The INVITE request is establishing the divert-to leg of a call,
2. The end-user device is authorized to use the diverting TN, based on the SHAKEN authentication criteria for applying full attestation defined in [shaken].

The criteria that the STI-AS applies for step-1 above to distinguish between an originating INVITE and a diverting INVITE will depend on the SIP profile supported by the end-user device and the host network, and specifically on how the end-user device and host network interwork to populate the header fields of the divert-to INVITE request. For example, in an IMS-compliant network where the P-CSCF asserts the identity of the initiating user in the P-Asserted-Identity header of an INVITE request received from an end-user device, the STI-AS could compare the TNs contained in the From and P-Asserted-Identity headers. If the TNs match, then the INVITE is establishing an originating call from the calling TN contained in the P-Asserted-Identity header. If the TNs don’t match, and the From TN is not assigned to the end-user device, then the INVITE is establishing the divert-to leg of a diverted call, where the From header identifies the originating TN, and the P-Asserted-Identity header identifies the diverting TN. Another more explicit indication of call diversion would be to use the presence and contents of a Diversion or History-Info header populated by the end-user device to record the diversion event.

If the header field contents indicate that an INVITE received from an end-user device is establishing an originating call, then the STI-AS shall perform the base SHAKEN authentication procedures defined in section 5.1. On the other hand, if the header field contents indicate that an INVITE request is establishing the divert-to leg of a diverted call, then the STI-AS shall perform two authentication procedures; the base SHAKEN authentication procedure for the calling TN as specified in section 5.1, and the "div" authentication procedure for the diverting TN as specified in section 5.3. In this case, the STI-AS shall assert an attestation level of "C" (Gateway) in the "shaken" PASSporT token, since the calling TN is not assigned to the diverting end-user device. Before sending the INVITE request to the next hop the STI-AS shall update the TN in the P-Asserted-Identity header to match the "orig" claim in the "shaken" Identity header.

*Open Issues:*

1. *Describe how this mechanism would support different call-forwarding cases; e.g., case-a) both originating TN-a and diverting TN-b are assigned to SIP-PBX, and case-b) call is forwarded multiple times before reaching SIP-PBX (these probably belong in Appendix).*
2. *Describe how this mechanism interworks with calling user privacy (Privacy:id). The host SP removes the PAID header before sending DID INVITE to SIP-PBX. Therefore, subsequent divert-to INVITE will have no info identifying calling TN-a. Presumably STI-AS can recognize this case by presence of Privacy:id header.*

#### Unable to determine INVITE-type

This section applies to the case where the end-user device does not provide sufficient information in INVITE requests to enable the STI-AS to distinguish between an originating and diverting INVVITE request.

On receiving an INVITE request from the end-user device, the STI-AS shall perform the base SHAKEN authentication procedures as specified in section 5.1. Since the STI-AS has no way of knowing whether the received INVITE request is for an originating or diverting call, it may *(shall?*) assert an attestation level of "C" (Gateway) for all calls.

# Appendix A

This Appendix provides background information the "div" PASSporT extension, and how it can enable end-to-end authentication of diverted calls.

Section 6.1 provides an overview of why SHAKEN needs an extension to support diverted calls.

Section 6.2describes how "div" PASSporT enables end-to-end call authentication when calls are diverted by a trusted in-network entity.

Section 6.3 describes how "div" PASSporT enables end-to-end call authentication for calls diverted by an untrusted end-user device.

## Overview of diverted calls and the impact to SHAKEN end-to-end call authentication

Based on the call authentication principles of SHAKEN, and specifically of PASSporT, in order for end-to-end call authentication to work properly, the SHAKEN authentication service in the originating network must both assert the telephone number identifying the originator of the call in the PASSporT "orig" claim, and specify the telephone number identifying the destination of the call in the PASSporT "dest" claim. The destination TN is included in the PASSporT token to provide protection from replay attacks where a man-in-the-middle replays a valid Identity header in a new INVITE sent to a different destination. In addition, PASSporT contains an "iat" claim that specifies the timestamp that the call was originated. Including the "iat” claim further limits the time associated with a potential replay of the specific "orig" and "dest" claims, to prevent a potential malicious flood of validated calls to the same destination TN.

A common SIP application functionality is to receive a call at the intended destination telephone number, and then retarget that call to another destination telephone number. This is generally referred to as call diversion, and is often used as part of a call-forwarding feature in a VoIP provider’s SIP application server or an end user’s SIP-PBX. Call diversion is also utilized by a number of other call scenarios in today’s telephone network; e.g., for features such as simultaneous ringing, where a call to the dialed TN is simultaneously offered to additional TN(s), and for toll-free number routing, where the dialed toll-free TN is replaced with its assigned routing TN.

Because call diversion inserts a new destination telephone number into the SIP INVITE without the explicit knowledge of the original calling party, this breaks the end-to-end call authentication model of SHAKEN/STIR. As a result, if not properly addressed, you could potentially have issues where a malicious entity may be able to utilize a call forwarding service to either inherit a valid call authentication or a valid entity may have the end-to-end call authentication broken because the original call authentication wasn’t applied to the resulting forwarded-to destination. Therefore, a specific solution is necessary to address the changed destination identity when a call is diverted.

To illustrate one of the potential vulnerabilities diversion presents into the SHAKEN call authentication framework, Figure 1 illustrates a man-in-the-middle replay attack window for the case where a malicious entity masquerades as another user by constructing an INVITE request that uses an Identity header from a legitimate call. The result of not specifying how diverted calls must be authenticated for every leg of the call enables the scenario illustrated in Figure 1, where the malicious call shown on the bottom of the figure is identical and indistinguishable from the legitimately forwarded/diverted call shown in the top of the figure.

 

Figure 1. Replayed INVITE looks like a legitimately diverted INVITE

Specifically, Figure 1 shows the initial INVITE message flow for a call from TN-a to TN-b that is forwarded to TN-c, where TN-a, TN-b and TN-c are served by SP-a, SP-b, and SP-c respectively. A malicious entity replays the To, P-Asserted-Identity and Identity headers in a new INVITE to called TN-x. Assuming the malicious user inserts a History-Info header corresponding to a “fake” forwarded call from TN-b to TN-x, the SHAKEN verification service is unable to distinguish between the legitimate [2] INVITE and the replayed [4] INVITE; they both pass the SHAKEN-defined verification tests. The fundamental issue is that the INVITE does not contain a PASSporT that proves the call was legitimately diverted from the original called TN identified in the To header field to a new destination TN identified in the Request-URI.

Corresponding to this scenario, a SHAKEN verification service that detects a valid SHAKEN PASSporT token where the PASSporT "dest" claim does not match the Request-URI TN, could result in interpretation of one of the following treatments:

* Treat as "verification passed” and incur the risk of providing a false-positive result, where a maliciously replayed INVITE is presented as "verification passed" to the called user.
* Treat as "verification failed” and incur the risk of providing a false-negative result, where a legitimately forwarded INVITE is presented as "verification failed" to the called user.

However, in order to avoid delivering a false-positive or false-negative indication to the called user, a baseline SHAKEN verification service must apply the flowing treatment:

* Treat as if no Identity header was received. The result is that SHAKEN is unable to deliver "verification passed" when an INVITE is legitimately forwarded or “verification failed” when an INVITE is maliciously retargeted within the "iat" freshness window, since neither condition can be determined or proven.

Call diversion, while not used in a large percentage of overall calls on a telephone network, is still significant and important for many telephone features, and ultimately needs to be addressed in the SHAKEN framework. The verification service behavior described above can be used temporarily for initial deployments of SHAKEN, with the goal of this document being to define the required extensions to SHAKEN to address end-to-end call authentication for call diversion.

There are two categories of call diversion applications that must be addressed. First is the case of in-network call diversion, where an application server that sits within a secure network domain applies call diversion. Second is the case end-user device call diversion, where an end user device such as a SIP-PBX, which is potentially in an untrusted domain, diverts a call back to the service provider network. The following two sections describe both of these categories of call diversion and how they apply to SHAKEN.

## SHAKEN support of "div" PASSporT for in-network call diversion

This section describes call authentication that would happen inside the trust domain of a service provider network. This general category of call features applied in the network corresponds to a relatively straightforward implementation of the “div” PASSporT extension to the SHAKEN framework.

As defined in draft-ietf-stir-passport-div, the “div” PASSporT extension defines a mechanism to provide a new PASSporT token at the point in the call where a diversion occurs.

Specific to SHAKEN, the first leg of the call is authenticated using the baseline SHAKEN Identity header containing a "shaken" PASSporT extension defined in draft-ietf-stir-passport-shaken as mandated by the SHAKEN framework. Each time a call is diverted, an additional Identity header will be added to the INVITE that follows the "div" PASSporT extension. Therefore, for diverted calls, there will be a single SHAKEN-based Identity header corresponding to the first leg of the call, and one or more "div" extension based Identity headers corresponding to each additional leg of the diverted call.

A "div" PASSporT token implicitly carries with it an attestation level of "Full Attestation” (using the “Full Attestation” semantics defined by SHAKEN). Therefore, an SP that adds a "div" Identity header is asserting with the strength indicated by "Full Attestation” that the diverting/forwarding TN is authorized to forward the call to the forward-to TN.

Figure 2 shows how the issue described in Figure 1 is resolved by the "div" PASSporT extension.



Figure 2. "div" PASSporT enables end-to-end authentication of legitimately diverted calls

As in Figure 1, Figure 2 shows the initial INVITE message sequence for a call from TN-a to TN-b that is forwarded to TN-c. Before forwarding the call, SP-b shall add a PASSporT "div" token to [2] INVITE to provide cryptographic proof that the call is being legitimately forwarded from TN-b to TN-c. Meanwhile, a malicious entity attempts to masquerade as TN-a by replaying the To, P-Asserted-Identity, Date and Identity headers from [2] INVITE into a new [4] INVITE to TN-x. The SHAKEN verification services in SP-c and SP-x can now distinguish between the legitimate call-forwarded call and the malicious call by verifying that the "shaken" and "div" PASSporT tokens provide an unbroken chain of authority between the final called TN identified in the Request-URI and the initial dialed TN identified in the "dest" claim of the "shaken" PASSporT token. In this example, the SHAKEN verification service in SP-x detects that the chain in [4] INVITE is broken, since the Request-URI TN does not match the “dest” claim in the “div” PASSporT token. As a result, SP-x includes a “fraud alert” indication in the [5] INVITE request to called UE-x.

An INVITE that is forwarded multiple times would have multiple "div" PASSporT tokens; one for each forwarded leg. The verification service that receives such an INVITE shall arrange the "div" PASSporT tokens in order, and verify the chain of authority from the Request-URI TN, through the multiple "div" PASSporT tokens to the "dest" TN in the "shaken" PASSporT token. Figure 3 illustrates the verification process for a multi-forwarding case, where TN-a calls TN-b, and the call is forward twice; first to TN-c, and then to TN-d.



 Figure 3. Multiple Diversion Example: TN-a calls TN-b fwd🡪 TN-c fwd🡪 TN-d

## SHAKEN support of "div" PASSporT for end-user device call diversion

This section describes how call authentication is supported when a call is diverted by an end-user device. (End-user devices that apply call diversion are typically SIP-PBX devices; therefore, we will refer to these devices using this term going forward.) A call leg diverted by a SIP-PBX can be authenticated using the "div" PASSporT extension, similar to the in-network call diversion cases described in section 4.2. If we were to apply the in-network procedures exactly, the diverting SIP-PBX would be responsible for adding the "div" PASSporT token. However, in order to speed up adoption, it would be preferable to have a solution that did not require the SIP-PBX to support "div" PASSporT. Therefore, the solution is designed such that host SP adds the "div" PASSporT token on behalf of the diverting SIP-PBX.

### SHAKEN functional requirements for call diverted by SIP-PBX

When a SIP-PBX applies call diversion, or more generally forwards a DID call, the calling number should be delivered to the forward-to service provider network. This calling number should correspond to the Identity header added by the SHAKEN authentication service in the originating network. For example, when a call is forwarded from a user’s office number to the user’s mobile number, the mobile phone should display both the actual calling number, and an accurate indication of the legitimacy of the calling number based on the SHAKEN verification results.

### Call-Forwarding Procedures

While there are many flavors of SIP-PBX devices that may act in slightly different ways, call diversion features are generally supported using similar procedures. The current prominent industry standard for SIP interworking between the SIP-PBX and its host service provider is SIPconnect 2.0. This document will therefore use SIPconnect 2.0 as the primary reference for describing how SHAKEN is supported for calls diverted by a SIP-PBX.

SIPconnect 2.0 defines two call-forwarding procedures for DID calls; the SIP-PBX can forward the call either by responding to the incoming INVITE request with a 302 Moved Temporarily response that redirects the call to the forward-to number, or by sending a new INVITE request to the forward-to number.

### Adding "div" PASSporT when SIP-PBX diverts call via 3xx Response

Figure 4 illustrates how "div" PASSporT can be utilized to provide end-to-end call authentication when a SIP-PBX diverts a DID call by sending a 302 Moved Temporarily response to the host SP to establish the forwarding leg of the call.



Figure 4. Support of "div" PASSporT when SIP-PBX diverts call with 3xx-response

**Figure 4 Message Sequence:**

1. SP-a performs SHAKEN authentication services for calling TN-a, and routes the call to SP-b.
2. On receiving [1] INVITE, SP-b verifies the SHAKEN PASSporT token, populates the P-Asserted-Identity header Verstat parameter with the "verification passed" result, and sends [2] INVITE to SIP-PBX-1.
3. SIP-PBX-1 responds to received [2] INVITE with [3] 302 Moved Temporarily response. The Contact header of the response contains the forward-to TN-c.
4. On receiving the 302-response, SP-b adds a 2nd Identity header to [4] INVITE containing a "div" PASSporT token that authenticates the forwarding leg of the call, and sends the [4] INVITE to SP-c.
5. On receiving [4] INVITE, SP-b verifies the received SHAKEN and "div" PASSporT tokens, populates the P-Asserted-Identity header Verstat parameter with the “verification passed” result, and sends [5] INVITE to UE-c. UE-c displays calling TN-a and a “verification passed” indication to the called user.

In order to add the "div" PASSporT token in step-4, SP-b must be authoritative for the forwarding TN identified in the "div" claim of the token (TN-b in this example). Fortunately, this will always be the case when a SIP-PBX diverts calls using a 3xx-response, since the SP that handles the 3xx-response is the same SP that routed the initial call leg to the SIP-PBX (i.e., by definition, the forwarding SP knows that the forwarding TN has been assigned to the SIP-PBX). This means that the message sequence shown in Figure 4 will support complex TN assignment cases where the SIP-PBX obtains TNs from multiple TN providers; e.g., SIP-PBX is assigned TNs from multiple host SPs, or toll-free TNs from a RespOrg, etc.

### Adding "div" PASSporT when SIP-PBX diverts call via new-INVITE Request

In this case, the host SP authenticates the forwarding (diverting) leg of a SIP-PBX call by adding an Identity header containing a "div" PASSporT token to the new-INVITE request that establishes the forward-to call leg. The host SP can do this only if it knows that the SIP-PBX is authorized to use the diverting TN. The host SP can establish the authority of the forwarding TN based on the STI authentication criteria for asserting full attestation defined in the base SHAKEN specification [ref].

The full attestation criteria defined by [shaken] enables a host SP to provide SHAKEN authentication with full attestation for DOD calls originated by a SIP-PBX; i.e., the host SP is able to verify with a high level of certainty that the originating SIP-PBX is authorized to originate calls from the calling TN. The following subsections describe how the full attestation criteria can be used by a host SP to verify the legitimacy of a SIP-PBX forwarding TN, so that the SP can add a “div” PASSporT token to a forwarding INVITE request received from a SIP-PBX.

#### Forwarding TN Authority based on SHAKEN Full Attestation criteria

Figure 5 shows the procedure for providing end-to-end call authentication when the SIP-PBX diverts a call by sending a new-INVITE to establish the forward-to leg, and the host SP determines the legitimacy of the forwarding TN based on the base SHAKEN authentication criteria for asserting full attestation; i.e., as specified by SHAKEN, the host Service Provider:

* Is responsible for the origination of the call onto the IP based service provider voice network.
* Has a direct authenticated relationship with the customer and can identify the customer.
* Has established a verified association with the telephone number used for the call.



Figure 5. Support for PBX divert via new-INVITE; forwarding TN authority based on SHAKEN full-attestation criteria

In addition to the fact that the call is diverted with a new INVITE request instead of a 302-response, the message sequence for Figure 5 differs from that shown in Figure 4 in two important ways:

1. SP-b knows that SIP-PBX-1 may divert DID calls using a new INVITE request. Therefore, in order to enable end-to-end delivery of the SHAKEN authentication information, SP-b includes the SHAKEN Identity header from [1] INVITE in the [2] INVITE request sent to SIP-PBX-1. SIP-PBX-1 in turn relays this Identity header in [3] INVITE back to SP-b so that it can be delivered to the forward-to network SP-c.
2. SP-b must implement logic to recognize that the [3] INVITE request received from SIP-PBX-1 is not establishing the initial leg of a DOD call, but is in fact establishing the forwarding leg of a previous DID call sent to the SIP-PBX. Therefore, instead of performing the base SHAKEN authentication procedures as it would for a normal DOD originating call, SP-b adds a "div" PASSporT token to authenticate the forwarding leg of the diverted call.

SP-b can assume that the received INVITE is legitimately establishing the forwarding leg of a diverted call if the following three conditions exist:

1. The received INVITE request contains a valid SHAKEN Identity header, and
2. The SIP-PBX is authorized to use the forwarding TN (based on SHAKEN full-attestation criteria).