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J. Rosenberg  
Cisco Systems  
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Obtaining and Using Globally Routable User Agent (UA) URIs (GRUU) in the  
Session Initiation Protocol (SIP)

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Abstract

Several applications of the Session Initiation Protocol (SIP) require a user agent (UA) to construct and distribute a URI that can be used by anyone on the Internet to route a call to that specific UA instance. A URI that routes to a specific UA instance is called a Globally Routable UA URI (GRUU). This document describes an extension to SIP for obtaining a GRUU from a server and for communicating a GRUU to a peer within a dialog.

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## 1. Introduction

The Session Initiation Protocol, [RFC 3261](#) [1], is used to establish and maintain a dialog between a pair of user agents in order to manage a communications session. Messages within the dialog are sent from one user agent to another using a series of proxy hops called the route set. They are eventually delivered to the remote target (the user agent on the other side of the dialog). This remote target is identified by a SIP URI obtained from the value of the Contact header field in INVITE requests and responses.

[RFC 3261](#) mandates that a user agent populate the Contact header field in INVITE requests and responses with a URI that is global (meaning that it can be used from any element connected to the Internet) and that routes to the user agent which inserted it. [RFC 3261](#) also mandates that this URI be valid for requests sent outside of the dialog in which the Contact URI was inserted.

In practice, these requirements have proven very difficult to meet. Few endpoints have a hostname that is present in DNS. Many endpoints have an IP address that is private because the client is behind a NAT. Techniques like the Simple Traversal of UDP Through NAT (STUN) [13] can be used to obtain IP addresses on the public Internet. However, many firewalls will prohibit incoming SIP requests from reaching a client unless they first pass through a proxy sitting in the DMZ of the network. Thus, URIs using STUN-obtained IP addresses often do not work.

Because of these difficulties, most clients have actually been inserting URIs into the Contact header field of requests and responses with the form sip:<IP-address>. These have the property of routing to the client, but they are generally only reachable from the proxy to which the user is directly connected. This limitation does not prevent SIP calls to an Address-of-Record (AOR) from proceeding because the user's proxy can usually reach these private addresses, and the proxy itself is generally reachable over the public network. However, this issue has impacted the ability of several other SIP mechanisms and applications to work properly.

An example of such an application is call transfer [21], based on the REFER method [7]. Another application is the usage of endpoint-hosted conferences within the conferencing framework [14]. Both of these mechanisms require that the endpoint be able to construct a URI that not only routes to that user agent, but is usable by entities anywhere on the Internet as a target for new SIP requests.

This specification formally defines a type of URI called a Globally Routable User Agent URI (GRUU) which has the properties of routing to



the UA and being reachable from anywhere. Furthermore, it defines a new mechanism by which a client can obtain a GRUU from its SIP provider, allowing it to use that URI in the Contact header fields of its dialog-forming or target refresh requests and responses. Because the GRUU is provided by the user's SIP provider, the GRUU properties can be guaranteed by the provider. As a result, the various applications which require the GRUU property, including transfer, presence, and conferencing, can work reliably.

## **2. Terminology**

The key words "MUST", "MUST NOT", "REQUIRED", "SHALL", "SHALL NOT", "SHOULD", "SHOULD NOT", "RECOMMENDED", "MAY", and "OPTIONAL" in this document are to be interpreted as described in [RFC 2119](#) [5].

This specification defines the following additional terms:

**contact:** The term "contact", when used in all lowercase, refers to a URI that is bound to an AOR or GRUU by means of a registration. A contact is usually a SIP URI, and is bound to the AOR and GRUU through a REGISTER request by appearing as the value of the Contact header field.

**remote target:** The term "remote target" refers to a URI that a user agent uses to identify itself for receipt of both mid-dialog and out-of-dialog requests. A remote target is established by placing a URI in the Contact header field of a dialog-forming request or response and updated by target refresh requests.

**Contact header field:** The term "Contact header field", with a capitalized C, refers to the header field which can appear in REGISTER requests and responses, redirects, or in dialog-creating requests and responses. Depending on the semantics, the Contact header field sometimes conveys a contact, and sometimes conveys a remote target.

## **3. Defining a GRUU**

URIs have properties, which are granted to the URI based on the policies of the domain that owns the URI. Those properties are not necessarily visible by inspection of the URI. In this context, the domain that owns the URI is the one indicated in the host part of the SIP URI. Some of the properties that a domain can confer upon a URI are:



The AOR property: A URI has the Address of Record (AOR) property if a domain will allow it to appear in the To header field of REGISTER request.

The alias property: A URI is an alias if its treatment by the domain is identical to another URI.

The service treatment property: A URI has the service treatment property if the domain will apply applications, features, and services to calls made by, or made to, that URI, possibly based on associating that URI with a user that has "subscribed" to various features.

The anonymous property: A URI has the anonymous property when it is not possible, by inspection of the URI, to discern the user with whom the URI is associated.

The identity property: A URI is considered an identity when the domain will authorize it as a valid value in the From header field of a request, such that an authentication service will sign a request with that URI [[16](#)].

This specification focuses on a property, called the Globally Routable User Agent URI (GRUU) property. A URI possesses this property when the following is true:

Global: It can be used by any User Agent Client (UAC) connected to the Internet. In that regard, it is like the address-of-record (AOR) property. A URI with the AOR property (for example, sip:joe@example.com), is meant to be used by anyone to reach that user. The same is true for a URI with the GRUU property.

Routes to a Single Instance: A request sent to that URI will be routed to a specific UA instance. In that regard, it is unlike the address-of-record property. When a request is sent to a URI with the AOR property, routing logic is applied in proxies to deliver the request to one or more UAs. That logic can result in a different routing decision based on the time of day, or the identity of the caller. However, when a request is made to a URI with the GRUU property, the routing logic is dictated by the GRUU property. The request has to be delivered to a very specific UA instance. That UA instance has to be the same UA instance for all requests sent to that URI.

Long Lived: The URI with the GRUU property persists for relatively long periods of time, ideally being valid for the duration of existence of the AOR itself. This property cannot be completely guaranteed, but providers are supposed to do their best to make





sure that a GRUU remains viable indefinitely.

A URI can have any combination of these properties. It is the responsibility of the domain which mints the URI to determine what properties are conferred upon that URI. This specification imposes requirements on a domain that mints a URI with the GRUU property.

For convenience, a URI that possesses the GRUU property is also referred to as a GRUU.

## **4. Use Cases**

There are several use cases where the GRUU properties are truly needed in order for a SIP application to operate.

### **4.1. REFER**

Consider a blind transfer application [21]. User A is talking to user B. User A wants to transfer the call to user C. So, user A sends a REFER to user C. That REFER looks like, in part:

```
REFER sip:C@example.com SIP/2.0
From: sip:A@example.com;tag=99asd
To: sip:C@example.com
Refer-To: (URI that identifies B's UA)
```

The Refer-To header field needs to contain a URI that can be used by user C to place a call to user B. However, this call needs to route to the specific UA instance that user B is using to talk to user A. If it doesn't, the transfer service will not execute properly. This URI is provided to user A by user B. Because user B doesn't know who user A will transfer the call to, the URI has to be usable by anyone. Therefore, it needs to be a GRUU.

### **4.2. Conferencing**

A similar need arises in conferencing [14]. In that framework, a conference is described by a URI that identifies the focus of the conference. The focus is a SIP UA that acts as the signaling hub for the conference. Each conference participant has a dialog with the focus. One case described in the framework is where a user A has made a call to user B. User A puts user B on hold, and calls user C. Now, user A has two separate dialogs for two separate calls -- one to user B, and one to user C. User A would like to conference them. To do this, user A's user agent morphs itself into a focus. It sends a re-INVITE or UPDATE [4] on both dialogs, and provides user B and user C with an updated remote target that now holds the conference URI.



The URI in the Contact header field also has a callee capabilities [8] parameter which indicates that this URI is a conference URI. User A proceeds to mix the media streams received from user B and user C. This is called an ad-hoc conference.

At this point, normal conferencing features can be applied. That means that user B can send another user, user D, the conference URI, perhaps in an email. User D can send an INVITE to that URI, and join the conference. For this to work, the conference URI used by user A in its re-INVITE or UPDATE has to be usable by anyone, and it has to route to the specific UA instance of user A that is acting as the focus. If it doesn't, basic conferencing features will fail. Therefore, this URI has to be a GRUU.

#### **4.3. Presence**

In a SIP-based presence [23] system, the Presence Agent (PA) generates notifications about the state of a user. This state is represented with the Presence Information Document Format (PIDF) [20]. In a PIDF document, a user is represented by a series of tuples, each of which describes the services that the user has. Each tuple also has a URI in the <contact> element, which is a SIP URI representing that service. A watcher can make a call to that URI, with the expectation that the call is routed to the service whose presence is represented in the tuple.

In some cases, the service represented by a tuple may exist on only a single user agent associated with a user. In such a case, the URI in the presence document has to route to that specific UA instance. Furthermore, since the presence document could be used by anyone who subscribes to the user, the URI has to be usable by anyone. As a result, it has to be a GRUU.

It is interesting to note that the GRUU may need to be constructed by a presence agent, depending on how the presence document is computed by the server.

### **5. Overview of Operation**

This section is tutorial in nature, and does not specify any normative behavior.

This extension allows a UA to obtain a GRUU, and to use a GRUU. These two mechanisms are separate, in that a UA can obtain a GRUU in any way it likes, and use the mechanisms in this specification to use it. This specification defines two mechanisms for obtaining a GRUU -- through registrations and through administrative operation. Only



the former requires protocol operations.

A UA can obtain a GRUU by generating a normal REGISTER request, as specified in [RFC 3261](#) [1]. This request contains a Supported header field with the value "gruu", indicating to the registrar that the UA supports this extension. The UA includes a "+sip.instance" Contact header field parameter of each contact for which a GRUU is desired. This parameter, defined in [11], contains a globally unique ID that identifies the UA instance. If the domain that the user is registering against also supports GRUU, the REGISTER responses will contain the "gruu" parameter in each Contact header field. This parameter contains a URI which the domain guarantees will route to that UA instance. This URI is tagged as a GRUU through the inclusion of a "gruu" URI parameter, similar to the way loose route URIs are tagged with the "lr" URI parameter. This GRUU is associated with the UA instance. Should the client change its contact, but indicate that it represents the same instance ID, the server would provide the same GRUU. Furthermore, if the registration for the contact expires, and the UA registers the contact at a later time with the same instance identifier, the server would provide the same GRUU.

Since the GRUU is a URI like any other, it can be handed out by a UA by placing it in any header field which can contain a URI. A UA will place the GRUU into the Contact header field of dialog forming and target refresh requests and responses it generates. [RFC 3261](#) mandates that the Contact header field have the GRUU property, and this specification provides a reliable way for a UA to obtain one. In other words, clients use the GRUU as a remote target. However, since the remote target used by clients to date has typically not had the GRUU properties, implementations have adapted their behaviors (oftentimes in proprietary ways) to compensate. To facilitate a transition away from these behaviors, it is helpful for a UA receiving the message to know whether the remote target is a GRUU or not. This can be known to a remote target through the presence of the "gruu" URI parameter.

A domain can construct a GRUU in any way it chooses. However, it is sometimes desirable to construct GRUUs so that any entity that receives a GRUU can determine the AOR for the subscriber associated with the UA instance. To facilitate that, the GRUU can be constructed such that it is identical to the subscriber's AOR, but includes the "opaque" and "gruu" URI parameters. The "opaque" URI parameter provides a general facility to construct a URI (such as a GRUU or a voicemail inbox for a user) that is related to an AOR, so that any element can extract the AOR from the constructed URI by removing the "opaque" parameter. It is because of the desire to use "opaque" for reasons besides GRUU, that both a "gruu" flag and an "opaque" URI parameter are defined. For example:



AOR: sip:alice@example.com  
GRUU: sip:alice@example.com;opaque="kjh29x97us97d";gruu

When a proxy in the domain constructs the GRUU, it would set the value of the "opaque" URI parameter such that it includes the instance ID. As such, when that proxy receives a request sent to the GRUU, it can determine that the request is a GRUU by the presence of the "gruu" parameter, and then it can extract the AOR and instance ID, both of which are needed to process the request.

When a UA uses a GRUU that routes to itself, it has the option of adding the "grid" URI parameter to the GRUU. This parameter is opaque to the proxy server handling the domain. However, when the server maps the GRUU to the contact bound to it, the server will copy the "grid" parameter and its value into the registered contact, and use the result in the Request-URI. As a result, when the UA receives the request, the Request-URI will contain the "grid" parameter it placed in the corresponding GRUU. If the GRUU did not contain a "grid" URI parameter, the server will insert a "grid" parameter into the Request-URI, but with no value. This signals to the UA that the request was sent to a GRUU.

The "grid" and "opaque" URI parameters play similar roles, but complement each other. The "opaque" parameter is added by the owner of the domain to correlate the GRUU to its instance ID, and to easily recognize that the URI has the GRUU property. The "grid" parameter is added by the UA instance so that, when a request is received by that instance, it can determine the context of the request.

## **6. Creation of a GRUU**

A GRUU is a URI that is created and maintained by a server authoritative for the domain in which the GRUU resides. Independently of whether the GRUU is created as a result of a registration or some other means, a server maintains certain information associated with the GRUU. This information, and its relationship with the GRUU, is modeled in Figure 1.





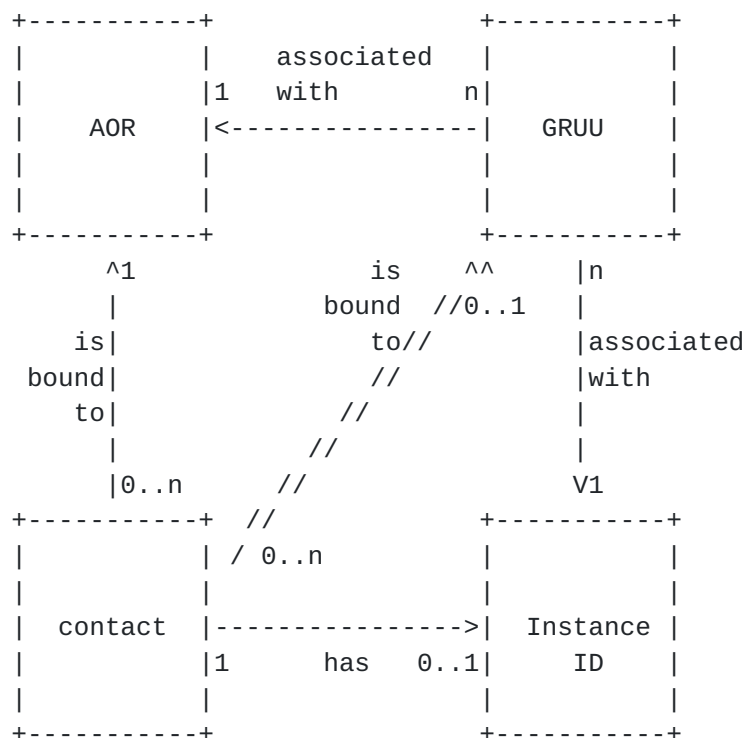


Figure 1

The instance ID plays a key role in this specification. The instance ID is defined in [11]. It is an identifier, represented with a URN, that uniquely identifies a SIP user agent amongst all other user agents associated with an AOR.

A GRUU is associated, in a many-to-one fashion, with the combination of an instance ID and an AOR. This combination is referred to as an instance ID/AOR pair. For each GRUU, there is one instance ID/AOR pair, and for each instance ID/AOR pair, there can be one or more GRUU. More than one GRUU might be defined in order to have aliases or URI that are anonymous or have other URI properties. However, this specification doesn't define any way for the client to learn about or use more than a single GRUU for each instance ID/AOR pair. The instance ID/AOR pair serves to uniquely identify a user agent instance servicing a specific AOR. The AOR identifies a resource, such as a user or service within a domain, and the instance ID identifies a specific UA instance servicing requests for that resource.

It is important to understand that GRUU is associated with the instance ID/AOR pair, not just the instance ID. For example, let's say a user registered the contact sip:ua@pc.example.com to the AOR sip:user@example.com, and included a +sip.instance="urn:foo:1"



parameter in the Contact header field. If the user also registered the contact sip:ua-112@pc.example.com with the same +sip.instance Contact header field parameter to a second AOR (say sip:boss@example.com), each of those UA instances would have a different GRUU because they belong to different AORs. That is the reason why a single instance ID can be associated with multiple GRUU; there would be one such association for each AOR. The same goes for the association of AOR to GRUU; there would be one such association for each instance ID.

The contacts that are bound to the GRUU are always the ones that have the instance ID that is associated with that GRUU. If none of the contacts bound to the AOR have the instance ID associated with the GRUU, then there are no contacts bound to the GRUU. If a contact should become registered to the AOR that has an instance ID equal to the one associated with the GRUU, that contact also becomes bound to the GRUU. If that contact should expire, it will no longer be bound to the AOR, and similarly, it will no longer be bound to the GRUU. The URI of the contact is irrelevant in determining whether it is bound to a particular GRUU; only the instance ID and AOR are important.

This specification does not mandate a particular mechanism for construction of the GRUU. Several example approaches are given in [Appendix A](#). However, the GRUU MUST exhibit the following properties:

- o The domain part of the URI is an IP address present on the public Internet, or, if it is a host name, the resolution procedures of [RFC 3263](#) [2], once applied, result in an IP address on the public Internet.
- o When a request is sent to the GRUU, it routes to a server that can make sure the request is delivered to the UA instance. For GRUUs created through registrations, this means that the GRUU has to route to a proxy server with access to registration data.
- o The URI MUST include the "gruu" URI parameter.
- o For each GRUU, both the SIP and Session Initiation Protocol Secure (SIPS) versions MUST exist. The SIPS URI may not always work, particularly if the proxy cannot establish a secure connection to the client. However, the SIPS URI always exists.
- o If the GRUU contains an "opaque" URI parameter, the URI that results from stripping out the "opaque" URI parameter MUST be equivalent to the AOR associated with the GRUU.

[Section 8.2](#) defines additional behaviors that a proxy must exhibit on



receipt of a GRUU.

When a domain constructs a URI with the GRUU properties, it MAY confer other properties upon this URI as a matter of domain policy. A domain can elect to confer properties like identity, anonymity, and service treatment. There is nothing in this specification that can allow the recipient of the GRUU to determine which of these properties (besides the GRUU property itself) have been conferred to the URI.

The service treatment property merits further discussion. Typically, the services a proxy executes upon receipt of a request sent to a GRUU will be a subset of those executed when a request is sent to the AOR. For requests that are outside of a dialog, it is RECOMMENDED to apply screening types of functions, both automated (such as black and white list screening) and interactive (such as interactive voice response (IVR) applications that confer with the user to determine whether to accept a call). In many cases, the new request is related to an existing dialog, and may be an attempt to join it (using the Join header field [24]) or replace it (using the Replaces header field [25]). In such cases, the UA will typically make its own authorization decisions, allowing the request if the sender can prove it knows the dialog identifiers [15]. In such cases, bypassing screening services might make sense, but it needs to be carefully considered by network designers, as it depends on the specific type of screening service.

However, forwarding services, such as call forwarding, SHOULD NOT be provided for requests sent to a GRUU. The intent of the GRUU is to target a specific UA instance, and this is incompatible with forwarding operations.

Mid-dialog requests will also be sent to GRUUs, as they are included as the remote-target in dialog-forming and target refresh requests and responses. However, in those cases, a proxy SHOULD only apply services that are meaningful for mid-dialog requests, generally speaking. This excludes screening functions, as well as forwarding ones.

The "opaque" URI parameter, defined in [Section 9](#), provides a means for a domain to construct a GRUU such that the AOR associated with the GRUU is readily extractable from the GRUU. Unless the GRUU is meant to also possess the anonymity property, it is RECOMMENDED that GRUUs be constructed using this parameter.

Because the GRUU is associated with both the instance ID and AOR, for any particular AOR there can be a potentially infinite number of GRUUs, and potentially more than one for each instance ID. However,



the instance IDs are only known to the network when an instance actually registers with one. As a result, it is RECOMMENDED that a GRUU be created at the time a contact with an instance ID is first registered to an AOR (even if that registration indicates that the registering UA doesn't even support GRUUs), until the time that the AOR is no longer valid in the domain. In this context, the GRUU exists if the domain, upon receiving a request for that GRUU, recognizes it as a GRUU, can determine the AOR and instance ID associated with it, and translate the GRUU to a contact if there is one with that instance ID currently registered. This property of the GRUU (existing from the time the first registration until removal of the AOR) can be difficult to achieve through software failures and power outages within a network, and for this reason, providing the property is at RECOMMENDED strength, and not MUST.

## **7. Obtaining a GRUU**

A GRUU can be obtained in many ways. This document defines two -- through registrations and through administrative operation.

### **7.1. Through Registrations**

When a GRUU is associated with a user agent that comes and goes, and registers itself to the network to bind a contact to an AOR, a GRUU is provided to the user agent through SIP REGISTER messages.

#### **7.1.1. User Agent Behavior**

##### **7.1.1.1. Generating a REGISTER Request**

When a UA compliant to this specification generates a REGISTER request (initial or refresh), it MUST include the Supported header field in the request. The value of that header field MUST include "gruu" as one of the option tags. This alerts the registrar for the domain that the UA supports the GRUU mechanism.

Furthermore, for each contact for which the UA desires to obtain a GRUU, the UA MUST include a "sip.instance" media feature tag [[11](#)] as a UA characteristic [[8](#)]. The instance ID MUST identify the UA that is performing the registration.

If a UA instance is registering against multiple AORs, it is RECOMMENDED that a UA instance provide a different contact URI for each AOR. This is needed for the UA to determine which GRUU to use as the remote target in responses to incoming dialog-forming requests, as discussed in [Section 8.1.1](#).





If a UA instance is trying to register multiple contacts for the same instance for the purposes of redundancy, it MUST use the procedures defined in [\[11\]](#).

Besides the procedures discussed above, the REGISTER request is constructed as it is in the case where this extension was not understood. Specifically, the contact in the REGISTER request SHOULD NOT contain the gruu Contact header field parameter, and the contact URI itself SHOULD NOT contain the "grid" URI parameter defined below. Any such parameters are ignored by the registrar, as the UA cannot propose a GRUU for usage with the contact. Typically, the contact URI will not contain the "gruu" URI parameter. However, it is permitted for a UA to register any kind of contact it wishes, including a GRUU. A typical use case for registering a GRUU is if a user wanted calls to one AOR to be delivered to a specific UA instance associated with another AOR. If a client does include a contact which contains a GRUU, that GRUU MUST NOT be associated with the AOR in the To header field of the REGISTER request. Doing so would result in a recursive definition for the GRUU and likely cause an infinite loop.

A UA MAY perform third party registrations (registrations where the entity performing the registration is not the same as the AOR in the To header field of the registration), and as in the above paragraph, MAY register contacts that do not point to the UA performing the registration. In addition, a UA MAY register contacts which omit the "+sip.instance" Contact header field parameter, in which case they would not be associated with any GRUU.

If a UA wishes to guarantee that the request is not processed unless the domain supports and uses this extension, it MAY include a Require header field in the request with a value that contains the "gruu" option tag. This is in addition to the presence of the Supported header field.

#### **[7.1.1.2.](#) Processing the REGISTER Response**

If the response is a 2xx, each Contact header field that contained the "+sip.instance" Contact header field parameter may also contain a "gruu" Contact header field parameter (which is distinct from the "gruu" URI parameter). This parameter contains a SIP or SIPS URI that represents a GRUU corresponding to the UA instance that registered the contact. The URI will be a SIP URI if the To header field in the REGISTER request contained a SIP URI, else (if the To header field in the REGISTER request contained a SIPS URI) it will be a SIPS URI. Any requests sent to the GRUU URI will be routed by the domain to a contact with that instance ID. Normally, the GRUU will not change in subsequent 2xx responses to REGISTER. Indeed, even if



the UA lets the contact expire, when it re-registers it at any later time, the registrar will normally provide the same GRUU for the same address-of-record and instance ID. However, as discussed above, this property cannot be completely guaranteed, as network failures may make it impossible to provide an identifier that persists for all time. As a result, a UA **MUST** be prepared to receive a different GRUU for the same instance ID/AOR pair in a subsequent registration response.

A non-2xx response to the REGISTER request has no impact on any existing GRUU previously provided to the UA. Specifically, if a previously successful REGISTER request provided the UA with a GRUU, a subsequent failed request does not remove, delete, or otherwise invalidate the GRUU.

#### **7.1.2. Registrar Behavior**

A registrar **MAY** create a GRUU for a particular instance ID/AOR pair at any time. Of course, if a UA requests a GRUU in a registration, and the registrar has not yet created one, it will need to do so in order to respond to the registration request. However, the registrar can create the GRUU in advance of any request from a UA.

A registrar **MUST** create both the SIP and SIPS versions of the GRUU, such that if the GRUU exists, both URI exist.

##### **7.1.2.1. Processing a REGISTER Request**

A REGISTER request might contain a Require header field; this indicates that the registration has to understand this extension in order to process the request.

As the registrar is processing the contacts in the REGISTER request according to the procedures of step 7 in [Section 10.3 of RFC 3261](#), the registrar checks whether each Contact header field in the REGISTER message contains a "+sip.instance" header field parameter. If present, the contact is processed further. If the registrar had not yet created a GRUU for that instance ID/AOR pair, it **MUST** do so at this time according to the procedures of [Section 6](#). If the contact contained a "gruu" Contact header field parameter, it **MUST** be ignored by the registrar. A UA cannot suggest or otherwise provide a GRUU to the registrar.

Next, the registrar checks if the contact URI is itself a GRUU associated with the AOR in the To header field of the registration. If it is, the registrar **MUST** reject the request with a 403 (Forbidden) error response.



Registration processing then continues as defined in [RFC 3261](#). If, after that processing, that contact is bound to the AOR, it also becomes bound to the GRUU associated with that instance ID/AOR pair. If, after that processing, the contact was not bound to the AOR (due, for example, to an expiration of zero), the contact is not bound to the GRUU either.

When generating the 200 (OK) response to the REGISTER request, the procedures of step 8 of [Section 10.3 of RFC 3261](#) are followed. Furthermore, for each Contact header field value placed in the response, if the registrar has stored an instance ID associated with that contact, that instance ID is returned as a Contact header field parameter. If the REGISTER request contained a Supported header field that included the "gruu" option tag, the server MUST add a "gruu" Contact header field parameter to that Contact header field. The value of the gruu parameter is a quoted string containing the URI that is the GRUU for the associated instance ID/AOR pair. If the To header field in the REGISTER request contains a SIP URI, the SIP version of the GRUU is returned. If the To header field in the REGISTER request contains a SIPS URI, the SIPS version of the GRUU is returned.

Note that handling of a REGISTER request containing a Contact header field with value "\*" and an expiration of 0 still retains the meaning defined in [RFC 3261](#) -- all contacts, not just those with a specific instance ID, are deleted. This removes the binding of each contact to the AOR and the binding of each contact to a GRUU.

Inclusion of a GRUU in the "gruu" Contact header field parameter of a REGISTER response is separate from the computation and storage of the GRUU. It is possible that the registrar has computed a GRUU for one UA, but a different UA that queries for the current set of registrations doesn't understand GRUU. In that case, the REGISTER response sent to that second UA would not contain the "gruu" Contact header field parameter, even though the UA has a GRUU for that contact. Similarly, a UA might send a REGISTER request with a contact containing a "+sip.instance" Contact header field but no "gruu" option tag in the Supported header field. The registrar can still assign a GRUU, and indeed, a subscriber to the registration event package could learn the GRUU from the notification [[26](#)] [[27](#)].

There is no need for inclusion of either a Require or Supported header field in the response with the "gruu" option tag.

#### **[7.1.2.2. Timing Out a Registration](#)**

When a registered contact expires, its binding to the AOR is removed as usual. In addition, its binding to the GRUU is removed at the



same time.

## **7.2. Through Administrative Operation**

Administrative creation of GRUUs is useful when a UA instance is a network server that is always available, and therefore doesn't register to the network. Examples of such servers are voicemail servers, application servers, and gateways.

There are no protocol operations required to administratively create a GRUU. The proxy serving the domain is configured with the GRUU, and with the contact to which it should be translated. It is not strictly necessary to also configure the instance ID and AOR, since the translation can be done directly. However, they serve as useful tools for determining to which resource and UA instance the GRUU is supposed to map.

In addition to configuring the GRUU and its associated contact in the proxy serving the domain, the GRUU will also need to be configured into the UA instance associated with the GRUU.

It is also reasonable to model certain network servers (such as PSTN gateways and media servers) as logically containing both a proxy and a UA instance. The proxy receives the request from the network, and passes it internally to the UA instance. In such a case, the GRUU routes directly to the server, and there is no need for a translation of the GRUU to a contact. The server itself would construct its own GRUU. When such a server constructs a GRUU, it MUST include the "gruu" URI parameter in it.

## **8. Using the GRUU**

### **8.1. UA Behavior**

#### **8.1.1. Sending a Message Containing a GRUU**

A UA first obtains a GRUU using the procedures of [Section 7](#), or by other means outside the scope of this specification.

A UA can use the GRUU in the same way it would use any other SIP or SIPS URI. However, a UA compliant to this specification SHOULD use a GRUU when populating the Contact header field of dialog-forming and target refresh requests and responses. In other words, a UA compliant to this specification SHOULD use its GRUU as its remote target. This includes the INVITE request, its 2xx response, the SUBSCRIBE [\[6\]](#) request, its 2xx response, the NOTIFY request, and the REFER [\[7\]](#) request, and its 2xx response.





If the UA instance has obtained multiple GRUUs for different AORs as a result of a registration, it SHOULD use one corresponding to the AOR used to send or receive the request. For sending a request, this means that the GRUU corresponds to the AOR present in the From header field. Furthermore, this means that the credentials used for authentication of the request correspond to the ones associated with that AOR. When receiving a request, the GRUU in the response corresponds to the AOR to which the original request was targeted. However, that AOR will be rewritten by the proxy to correspond to the UA's registered contact. It is for this reason that different contacts are needed for each AOR that an instance registers against. In this way, when an incoming request arrives, the Request URI can be examined. It will be equal to a registered contact. That contact can be used to map directly to the AOR, and from there, the correct GRUU can be selected.

When using a GRUU as a remote target, a UA MAY add the "grid" URI parameter to the GRUU, and if one is added, MUST include a value for it. This parameter MAY take any value permitted by the grammar for the parameter. When a UA sends a request to the GRUU, the proxy for the domain that owns the GRUU will translate the GRUU in the Request-URI, replacing it with the contact bound to that GRUU. However, the proxy will retain the "grid" parameter and its value when this translation is performed. As a result, when the UA receives the request, the Request-URI will contain the "grid" created by the UA. This allows the UA to effectively manufacture an infinite supply of GRUU, each of which differs by the value of the "grid" parameter. When a UA receives a request that was sent to the GRUU, it will be able to tell which GRUU was invoked by looking at the "grid" parameter.

An implication of this behavior is that all mid-dialog requests will be routed through intermediate proxies. There will never be direct, UA-to-UA signaling unless the UA is co-resident with the proxy (which is the case for administratively constructed GRUUs).

When a UA requires a URI with the GRUU properties in order to reach a peer for a particular SIP application (such as assisted call transfer), it uses the URI in the Contact header field of a request or response from that peer if it contains a GRUU. This is trivially determined by the presence of the "gruu" URI parameter.

As per [RFC 3261](#), a UA SHOULD include a Supported header with the option tag "gruu" in requests it generates.

#### **8.1.2. Sending a Message to a GRUU**

There is no new behavior associated with sending a request to a GRUU.



A GRUU is a URI like any other. When a UA receives a request or response, it knows that the remote target is a GRUU by the presence of the "gruu" URI parameter. The UA can take the GRUU, send a request to it, and then be sure that the request is delivered to the UA instance which sent the request or response.

If the GRUU contains the "opaque" URI parameter, a UA can obtain the AOR for the user by stripping the "opaque" and "gruu" URI parameters. The resulting URI is the AOR. If the GRUU does not have the "opaque" URI parameter, there is no mechanism defined for determining the AOR from the GRUU. Extraction of the AOR from the GRUU is useful for call logs and other accounting functions where it is desirable to know the user to whom the request was directed.

Because the instance ID is a callee capabilities parameter, a UA might be tempted to send a request to the AOR of a user, and include an Accept-Contact header field [19] that indicates a preference for routing the request to a UA with a specific instance ID. Although this would appear to have the same effect as sending a request to the GRUU, it does not. The caller preferences expressed in the Accept-Contact header field are just preferences. Its efficacy depends on a UA constructing an Accept-Contact header field that interacts with domain-processing logic for an AOR, to cause a request to route to a particular instance. Given the variability in routing logic in a domain (for example, time-based routing to only selected contacts), this doesn't work for many domain-routing policies. However, this specification does not forbid a client from attempting such a request, as there may be cases where the desired operation truly is a preferential routing request.

#### **8.1.3. Receiving a Request Sent to a GRUU**

When a User Agent Server (UAS) receives a request sent to its GRUU, the incoming request URI will be equal to the contact that was registered (through REGISTER or some other action) by that UA instance. If the user agent had previously handed out its GRUU with a "grid" parameter, the incoming Request-URI may contain that parameter. This indicates to the UAS that the request is being received as a result of a request sent by the UAC to that GRUU/grid combination. This specification makes no normative statements about when to use a "grid" parameter, or what to do when receiving a request made to a GRUU/grid combination. Generally, any differing behaviors are a matter of local policy. If the UA had not included a "grid" parameter in the GRUU, the incoming request will still have a "grid" parameter, but with no value.

It is important to note that, when a user agent receives a request, the presence or absence of the "grid" parameter will inform the user



agent whether the request was sent to the AOR or to the GRUU. If the parameter is absent, it means that the request was sent to the AOR. If present, it means that the request was sent to the GRUU. This is true regardless of whether the UA itself uses the "grid" parameter, since the home proxy will insert one into the Request-URI when receiving a request sent to a GRUU.

## **8.2. Proxy Behavior**

Proxy behavior is fully defined in [Section 16 of RFC 3261](#) [1]. GRUU processing impacts that processing in two places -- request targeting and record routing.

### **8.2.1. Request Targeting for Requests Outside of a Dialog**

When a proxy server receives a request, owns the domain in the Request-URI, and is supposed to access a Location Service in order to compute request targets (as specified in [Section 16.5 of RFC 3261](#) [1]), the proxy examines the Request-URI. If the Request-URI is an AOR against which there are multiple registered contacts with the same instance ID parameter (allowing multiple flows, as defined in [11]), the proxy MUST follow the procedures defined there and populate the target set so that there is never more than one contact at a time with a given instance ID.

If the Request-URI is within the domain of the proxy, and the URI contains the "gruu" URI parameter, but the URI does not refer to a GRUU known within the domain, the proxy rejects the request with a 404 (Not Found). If the Request-URI is within the domain of the proxy, contains a "gruu" URI parameter, and the GRUU is known within the domain and refers to a valid AOR within the domain, but the instance ID is unknown, the proxy SHOULD generate a 480 (Temporarily Unavailable).

Otherwise, handling of the GRUU proceeds as specified in [RFC 3261 Section 16](#). For GRUUs, the abstract location service described in [Section 16.5](#) is utilized, producing a set of zero or more contacts, each of which is associated with the same instance ID. If there are more than one contact with the same instance ID, and those contact were registered using the procedures of [11], those procedures are used to select one. Otherwise, the most recently updated contact is used. This produces zero or one contacts. The server MUST copy the "grid" parameter from the Request-URI (if present) into the new target URI obtained from the registered contact. If there was no "grid" parameter in the Request-URI, the proxy MUST insert a "grid" parameter into the new target URI obtained from the registered contact, and MUST omit a value for the parameter. Note that the "gruu" URI parameter is not copied. If the grid was already present



in the contact bound to the GRUU, it is overwritten in this process. If no contacts were bound to the GRUU, the lookup of the GRUU in the abstract location service will result in zero target URIs, eventually causing the proxy to reject the request with a 480 (Temporarily Unavailable) response.

If the contact was registered using a Path header field [3], then that Path is used to construct the route set for reaching the contact through the GRUU, as well as through the AOR, using the procedures specified in [RFC 3327](#) [3]. However, support for GRUU at a registrar does not require support for [RFC 3327](#).

A proxy MAY apply other processing to the request, such as execution of called party features, as discussed in [Section 6](#).

A request sent to a GRUU SHOULD NOT be redirected. In many instances, a GRUU is used by a UA in order to assist in the traversal of NATs and firewalls, and a redirection may prevent such a case from working.

### **[8.2.2](#). Record-Routing**

The proxy that accesses the location service (called the home proxy here) MUST record-route under two circumstances. Firstly, if the home proxy receives a dialog forming request from a UA in its own domain (an originating request) whose Contact header field contains a GRUU (indicated by the presence of the "gruu" URI parameter), the proxy MUST record-route. Secondly, if the home proxy receives a dialog-forming request targeted to an AOR or GRUU within the domain of the home proxy (a terminating request), and it translates the Request URI into a contact that is associated with an instance ID to which a GRUU has been assigned, the proxy MUST record-route. The URI placed into the record-route MUST cause the request to be routed to a proxy that can access the location service for that user.

### **[8.2.3](#). Request Targeting for Mid-Dialog Requests**

When a mid-dialog request is sent to a UA which used its GRUU as the remote target, this mid-dialog request will arrive at the home proxy. As a consequence of the record-routing procedures in [Section 8.2.2](#), this request will arrive with a Request-URI equal to the GRUU, and the topmost Route header field equal to the URI placed into the Record-Route previously.

Proxy processing of this request is nearly identical to that of [Section 8.2.1](#). The proxy MUST look up the GRUU in the location service, and translate it to the registered contacts. If, based on the procedures of [Section 8.2.1](#), this lookup fails or produces no





contacts, the request MUST be rejected as described there. If the lookup produces a single registered contact, that contact is placed into the Request-URI. As with requests outside of a dialog, the "grid" URI parameter is placed into the translated URI (having either been copied from the Request-URI or placed there with no value otherwise). If there are multiple registered contacts (which happens when the client registers multiple flows using [11]), the proxy chooses one arbitrarily. The actual one that is chosen is not relevant; the Request-URI will not be used by an edge proxy compliant to [11] to deliver the request to the UA.

Once the proxy finishes its processing, it will pop the topmost Route header field value. If there were additional Route header field values beyond the one pointing to the home proxy, these are not touched or modified in any way by the procedures defined here. Any Path values that may have been registered are not used. If there were no additional Route header field values beyond the one pointing to the home proxy, any Path values that were registered MUST be used as if this was a request sent outside of any existing dialog.

The request is then forwarded based on the rules in [RFC 3261](#). This will use any Route header field values if present, else will use the Request-URI. If the Request-URI is being used, the request gets delivered using the procedures of [11] if the contact was registered using those mechanisms.

## **9. The opaque SIP URI Parameter**

This specification defines a new SIP URI parameter, "opaque". The "opaque" URI parameter is used to construct a URI (called the derived URI) that is related to another URI (called the base URI, frequently an AOR) in some way. In this specification, the parameter is used to construct the GRUU (the derived URI) from the AOR (the base URI). However, there are many other applications outside of GRUU. It can be used, for example, to construct a URI for a voicemail inbox (the derived URI) from a subscriber's AOR (the base URI), or the URI for a video service advertised via presence [22] (the derived URI) from the subscriber's AOR (the base URI).

To construct a derived URI, the owner of the domain adds the "opaque" URI parameter to the base URI, resulting in the derived URI. In fact, these are the only semantics associated with the "opaque" URI parameter: a URI containing the parameter MUST be related to another URI, obtained by stripping the "opaque" URI parameter. Because the "opaque" URI parameter implies a relationship, any element (including those outside the domain that owns the URI) that receives a URI with the "opaque" URI parameter will know definitively that it is a



derived URI, and can strip it to obtain the base URI.

The value of the "opaque" URI parameter is not relevant to anyone except for the owner of the domain. It typically contains information needed by the owner of the domain to correctly process a request targeted to that URI according to the desired semantics of the URI. As such, the parameter is a form of cookie. In the case of a GRUU, the "opaque" URI parameter contains enough information for the owner of the domain to determine the instance ID. Since the structure of its value is not subject to standardization, it can only be interpreted by the same proxy or cluster of proxies that created the derived URI. For this reason, a proxy or cluster of proxies **MUST NOT** create a derived URI unless a request sent to the base URI (and consequently the derived URI) will be routed back to that same proxy or cluster of proxies without any upstream proxies requiring interpretation of the "opaque" URI parameter. Simply put, a request sent to a derived URI has to get back to the same proxy farm that created the derived URI.

The presence of the "opaque" URI parameter in a URI implies a relationship between that URI and its base URI. However, the nature of that relationship cannot be determined from inspection of the URI alone. In some cases, there may be no way to know the relationship outside of the domain that constructed the URI. In other cases, as with GRUU, the nature of the relationship can be determined from the URI. When any element receives a URI with the "gruu" URI parameter, and that URI contains the "opaque" URI parameter, the URI formed by stripping the "opaque" and "gruu" URI parameter is the AOR associated with the GRUU.

## **10. Grammar**

This specification defines one new Contact header field parameter ("gruu") by extending the grammar for "contact-params" as defined in [RFC 3261](#). It also defines three new SIP URI parameters ("grid", "gruu" and "opaque") by extending the grammar for "uri-parameter" as defined in [RFC 3261](#). The grammar for string-value is obtained from [\[8\]](#).

```
contact-params    =/ c-p-gruu
c-p-gruu          = "gruu" EQUAL LDQUOTE (SIP-URI / SIPS-URI) RDQUOTE
uri-parameter     =/ grid-param / opaque-param / gruu-param
grid-param        = "grid" ["=" pvalue]           ; defined in RFC3261
opaque-param      = "opaque=" pvalue               ; defined in RFC3261
gruu-param        = "gruu"
```



## **11. Requirements**

This specification was created in order to meet the following requirements:

REQ 1: When a UA invokes a GRUU, it MUST cause the request to be routed to the specific UA instance to which the GRUU refers.

REQ 2: It MUST be possible for a GRUU to be invoked from anywhere on the Internet, and still cause the request to be routed appropriately. That is, a GRUU MUST NOT be restricted to use within a specific addressing realm.

REQ 3: It MUST be possible for a GRUU to be constructed without requiring the network to store additional state.

REQ 4: It MUST be possible for a UA to obtain a multiplicity of GRUUs that each route to that UA instance. For example, this is needed to support ad-hoc conferencing where a UA instance needs a different URI for each conference it is hosting.

REQ 5: When a UA receives a request sent to a GRUU, it MUST be possible for the UA to know the GRUU that was used to invoke the request. This is necessary as a consequence of REQ 4.

REQ 6: It MUST be possible for a UA to add opaque content to a GRUU. This content is not interpreted or altered by the network, and is used only by the UA instance to whom the GRUU refers. This provides a basic cookie type of functionality, allowing a UA to build a GRUU with the state embedded.

REQ 7: It MUST be possible for a proxy to execute services and features on behalf of a UA instance represented by a GRUU. As an example, if a user has call blocking features, a proxy may want to apply those call blocking features to calls made to the GRUU, in addition to calls made to the user's AOR.

REQ 8: It MUST be possible for a UA in a dialog to inform its peer of its GRUU, and for the peer to know that the URI represents a GRUU. This is needed for the conferencing and dialog reuse applications of GRUUs, where the URIs are transferred within a dialog.

REQ 9: When transferring a GRUU per REQ 8, it MUST be possible for the UA receiving the GRUU to be assured of its integrity and authenticity.



REQ 10: It MUST be possible for a server that is authoritative for a domain to construct a GRUU which routes to a UA instance bound to an AOR in that domain. In other words, the proxy can construct a GRUU, too. This is needed for the presence application.

## **12. Example Call Flow**

The following call flow, shown in Figure 2, shows a basic registration and call setup, followed by a subscription directed to the GRUU. It then shows a failure of the callee, followed by a re-registration. The conventions of [\[18\]](#) are used to describe representation of long message lines.





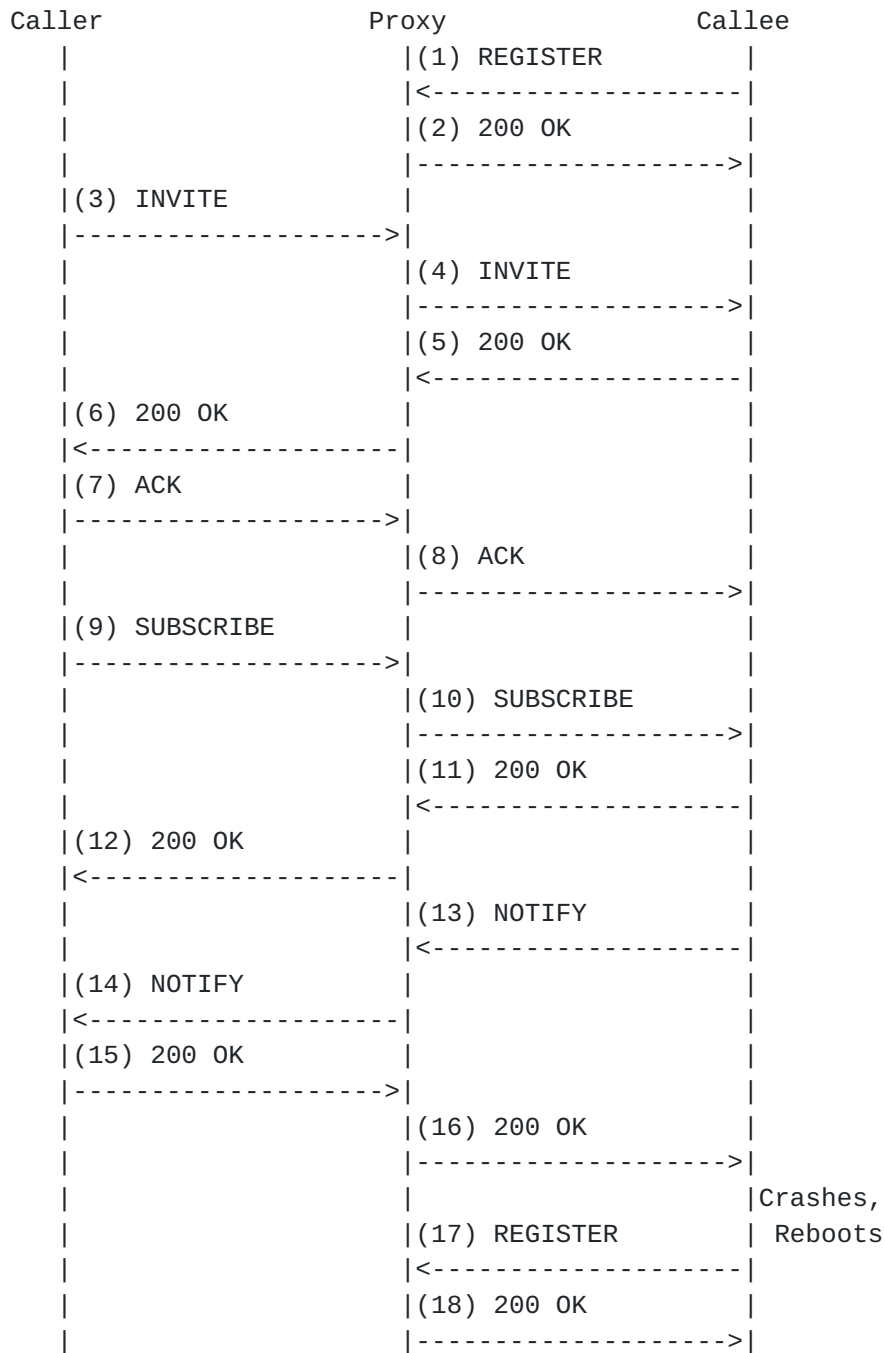


Figure 2

The Callee supports the GRUU extension. As such, its REGISTER (1) looks like:



```
REGISTER sip:example.com SIP/2.0
Via: SIP/2.0/UDP 192.0.2.1;branch=z9hG4bKnashds7
Max-Forwards: 70
From: Callee <sip:callee@example.com>;tag=a73kszlfl
Supported: gruu
To: Callee <sip:callee@example.com>
Call-ID: 1j9FpLxk3uxtm8tn@192.0.2.1
CSeq: 1 REGISTER
Contact: <sip:callee@192.0.2.1>
      ;+sip.instance="urn:uuid:f81d4fae-7dec-11d0-a765-00a0c91e6bf6>"
Content-Length: 0
```

The REGISTER response would look like:

```
SIP/2.0 200 OK
Via: SIP/2.0/UDP 192.0.2.1;branch=z9hG4bKnashds7
From: Callee <sip:callee@example.com>;tag=a73kszlfl
To: Callee <sip:callee@example.com> ;tag=b88sn
Call-ID: 1j9FpLxk3uxtm8tn@192.0.2.1
CSeq: 1 REGISTER
<allOneLine>
Contact: <sip:callee@192.0.2.1>
      ;gruu="sip:callee@example.com;gruu;
opaque=urn:uuid:f81d4fae-7dec-11d0-a765-00a0c91e6bf6"
      ;+sip.instance="urn:uuid:f81d4fae-7dec-11d0-a765-00a0c91e6bf6>"
      ;expires=3600
</allOneLine>
Content-Length: 0
```

Note how the Contact header field in the REGISTER response contains the gruu parameter with the URI sip:callee@example.com;gruu;opaque=urn:uuid:f81d4fae-7dec-11d0-a765-00a0c91e6bf6. This represents a GRUU that translates to the contact sip:callee@192.0.2.1.

The INVITE from the caller is a normal SIP INVITE. However, the 200 OK generated by the callee (message 5) now contains a GRUU as the remote target. The UA has also chosen to include a "grid" URI parameter into the GRUU.



```
SIP/2.0 200 OK
Via: SIP/2.0/UDP proxy.example.com;branch=z9hG4bKnaa8
Via: SIP/2.0/UDP host.example.com;branch=z9hG4bK99a
From: Caller <sip:caller@example.com>;tag=n88ah
To: Callee <sip:callee@example.com> ;tag=a0z8
Call-ID: 1j9FpLxk3uxtma7@host.example.com
CSeq: 1 INVITE
Supported: gruu
Allow: INVITE, OPTIONS, CANCEL, BYE, ACK, SUBSCRIBE
<allOneLine>
Contact:
<sip:callee@example.com;gruu
;opaque=urn:uuid:f81d4fae-7dec-11d0-a765-00a0c91e6bf6;grid=99a>
</allOneLine>
Content-Length: --
Content-Type: application/sdp

[SDP Not shown]
```

At some point later in the call, the caller decides to subscribe to the dialog event package [17] at that specific UA. To do that, it generates a SUBSCRIBE request (message 9), but directs it towards the remote target, which is a GRUU:

```
<allOneLine>
SUBSCRIBE sip:callee@example.com;gruu;opaque=urn:uuid:f8
1d4fae-7dec-11d0-a765-00a0c91e6bf6;grid=99a
SIP/2.0
</allOneLine>
Via: SIP/2.0/UDP host.example.com;branch=z9hG4bK9zz8
From: Caller <sip:caller@example.com>;tag=kkaz-
<allOneLine>
To: <sip:callee@example.com;opaque=urn:uuid:f8
1d4fae-7dec-11d0-a765-00a0c91e6bf6;grid=99a;gruu>
</allOneLine>
Call-ID: faif9a@host.example.com
CSeq: 2 SUBSCRIBE
Supported: gruu
Event: dialog
Allow: INVITE, OPTIONS, CANCEL, BYE, ACK, NOTIFY
Contact: <sip:caller@example.com;gruu;opaque=hdg7777ad7aflzig8sf7>
Content-Length: 0
```

In this example, the caller itself supports the GRUU extension, and is using its own GRUU to populate its remote target.

This request is routed to the proxy, which proceeds to perform a



location lookup on the Request-URI. It is translated into the contact for that instance, and then proxied to that contact. Note how the "grid" parameter is maintained, and the "gruu" parameter is no longer present.

```
SUBSCRIBE sip:callee@192.0.2.1;grid=99a SIP/2.0
Via: SIP/2.0/UDP proxy.example.com;branch=z9hG4bK9555
Via: SIP/2.0/UDP host.example.com;branch=z9hG4bK9zz8
From: Caller <sip:caller@example.com>;tag=kkaz-
<allOneLine>
To: <sip:callee@example.com;opaque=urn:uuid:f8
1d4fae-7dec-11d0-a765-00a0c91e6bf6;grid=99a;gruu>
</allOneLine>
Call-ID: faif9a@host.example.com
CSeq: 2 SUBSCRIBE
Supported: gruu
Event: dialog
Allow: INVITE, OPTIONS, CANCEL, BYE, ACK, NOTIFY
Contact: <sip:caller@example.com;opaque=hdg7777ad7aflzig8sf7>
Content-Length: 0
```

At some point after message 16 is received, the callee's machine crashes and recovers. It obtains a new IP address, 192.0.2.2. Unaware that it had previously had an active registration, it creates a new one (message 17 below). Notice how the instance ID remains the same, as it persists across reboot cycles:

```
REGISTER sip:example.com SIP/2.0
Via: SIP/2.0/UDP 192.0.2.2;branch=z9hG4bKnasbba
Max-Forwards: 70
From: Callee <sip:callee@example.com>;tag=ha8d777f0
Supported: gruu
To: Callee <sip:callee@example.com>
Call-ID: hf8asxzff8s7f@192.0.2.2
CSeq: 1 REGISTER
Contact: <sip:callee@192.0.2.2>
;+sip.instance="<urn:uuid:f81d4fae-7dec-11d0-a765-00a0c91e6bf6>"
Content-Length: 0
```

The registrar notices that a different contact, sip:callee@192.0.2.1, is already associated with the same instance ID. It registers the new one too and returns both in the REGISTER response. Both have the same GRUU. However, only this new contact (the most recently registered one) will be used by the proxy for population in the target set. The registrar then generates the following response:





```
SIP/2.0 200 OK
Via: SIP/2.0/UDP 192.0.2.2;branch=z9hG4bKnasbba
From: Callee <sip:callee@example.com>;tag=ha8d777f0
To: Callee <sip:callee@example.com>;tag=99f8f7
Call-ID: hf8asxzff8s7f@192.0.2.2
CSeq: 1 REGISTER
<allOneLine>
Contact: <sip:callee@192.0.2.2>
;gruu="sip:callee@example.com;gruu;opaque=urn:
uuid:f81d4fae-7dec-11d0-a765-00a0c91e6bf6"
;+sip.instance="<urn:uuid:f81d4fae-7dec-11d0-a765-00a0c91e6bf6>"
;expires=3600
</allOneLine>
<allOneLine>
Contact: <sip:callee@192.0.2.1>
;gruu="sip:callee@example.com;gruu;opaque=urn:
uuid:f81d4fae-7dec-11d0-a765-00a0c91e6bf6"
;+sip.instance="<urn:uuid:f81d4fae-7dec-11d0-a765-00a0c91e6bf6>"
;expires=400
</allOneLine>
Content-Length: 0
```

### 13. Security Considerations

It is important for a UA to be assured of the integrity of a GRUU given in a REGISTER response. If the GRUU is tampered with by an attacker, the result could be denial of service to the UA. As a result, it is RECOMMENDED that a UA use the SIPS URI scheme in the Request-URI when registering. Proxies and registrars MUST support the sips URI and MUST support TLS. Note that this does not represent a change from the requirements in [RFC 3261](#).

The example GRUU construction algorithm in [Appendix A.1](#) makes no attempt to create a GRUU that hides the AOR and instance ID associated with the GRUU. In general, determination of the AOR associated with a GRUU is considered a good property, since it allows for easy tracking of the target of a particular call. Learning the instance ID provides little benefit to an attacker. To register or otherwise impact registrations for the user, an attacker would need to obtain the credentials for the user. Knowing the instance ID is insufficient.

The example GRUU construction algorithm in [Appendix A.1](#) makes no attempt to create a GRUU that prevents users from guessing a GRUU based on knowledge of the AOR and instance ID. A user that is able to do that will be able to direct a new request at a particular instance. However, this specification recommends that service



treatment (in particular, screening features) be given to requests that are sent to a GRUU. That treatment will make sure that the GRUU does not provide a back door for attackers to contact a user that has tried to block the attacker.

GRUUs do not provide a solution for privacy. In particular, since the GRUU does not change during the lifetime of a registration, an attacker could correlate two calls as coming from the same source, which in and of itself reveals information about the caller. Furthermore, GRUUs do not address other aspects of privacy, such as the addresses used for media transport. For a discussion of how privacy services are provided in SIP, see [RFC 3323](#) [12].

As a consequence of this specification, a UA will begin using GRUU in the dialog forming and target refresh requests and responses it emits. These GRUU will be passed to other UA (called the correspondent), which then use them in requests that they emit. These UA might be malicious, and attempt to remove the "gruu", "grid", or "opaque" parameters from the URI before using it. Consequently, consideration must be given to the effect of such removal.

If a malicious correspondent removes the "gruu" URI parameter, the request will be routed to the home proxy. Despite the absence of the "gruu" parameter, the home proxy will still recognize the URI as a GRUU, based on the presence and value of the "opaque" parameter. Consequently, a home proxy SHOULD NOT rely solely on the presence of the "gruu" parameter to determine that a URI is a GRUU. If a malicious correspondent removes both the "gruu" and "opaque" URI parameters, the resulting URI will, in many cases, look identical to an AOR, and thus receive the same treatment as an AOR. If this is done in a mid-dialog request, the proxy might translate the AOR to a registered contact. If this registered contact points to a different UA instance than the one in the dialog, the request might be misrouted to that instance. Since the dialog doesn't exist there, the request is rejected. This has no harmful effects to anyone except for the malicious correspondent. If a malicious correspondent removes the "grid" parameter, the request will be delivered to the UA, but contain an empty "grid" parameter inserted by the home proxy. If a UA requires the "grid" parameter to process the request, then it SHOULD always insert a "grid" parameter into all GRUU it hands out, with a different value for each. Consequently, if a request should arrive with an empty "grid" parameter, the UA will know that the parameter had been stripped by a malicious correspondent, and it can reject the request if desired.



## **14. IANA Considerations**

This specification defines a new Contact header field parameter, three SIP URI parameters, and a SIP option tag.

### **14.1. Header Field Parameter**

This specification defines a new header field parameter, as per the registry created by [\[9\]](#). The required information is as follows:

Header field in which the parameter can appear: Contact

Name of the Parameter: gruu

RFC Reference: RFC XXXX [[NOTE TO IANA: Please replace XXXX with the RFC number of this specification.]]

### **14.2. URI Parameters**

This specification defines three new SIP URI parameters, as per the registry created by [\[10\]](#).

Name of the Parameter: grid

Predefined Values: none

RFC Reference: RFC XXXX [[NOTE TO IANA: Please replace XXXX with the RFC number of this specification.]]

Name of the Parameter: opaque

Predefined Values: none

RFC Reference: RFC XXXX [[NOTE TO IANA: Please replace XXXX with the RFC number of this specification.]]

Name of the Parameter: gruu

Predefined Values: none

RFC Reference: RFC XXXX [[NOTE TO IANA: Please replace XXXX with the RFC number of this specification.]]

### **14.3. SIP Option Tag**

This specification registers a new SIP option tag, as per the guidelines in [Section 27.1 of RFC 3261](#).



Name: gruu

Description: This option tag is used to identify the Globally Routable User Agent URI (GRUU) extension. When used in a Supported header, it indicates that a User Agent understands the extension. When used in a Require header field of a REGISTER request, it indicates that the registrar shouldn't process the registration unless it supports the GRUU extension.

## **15. Acknowledgements**

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## **[Appendix A.](#) Example GRUU Construction Algorithms**

The mechanism for constructing a GRUU is not subject to specification. This appendix provides two examples that can be used by a registrar. Of course, others are permitted, as long as they meet the constraints defined for a GRUU.

### **[A.1.](#) Instance ID in "opaque" URI Parameter**

The most basic approach for constructing a GRUU is to utilize the "opaque" URI parameter. The user and domain portions of the URI are equal to the AOR, and the "opaque" parameter is populated with the instance ID.

### **[A.2.](#) Encrypted Instance ID and AOR**

In many cases, it will be desirable to construct the GRUU in such a way that it will not be possible, based on inspection of the URI, to determine the Contact URI that the GRUU translates to. It may also be desirable to construct it so that it will not be possible to determine the instance ID/AOR pair associated with the GRUU. Whether a GRUU should be constructed with this property is a local policy



decision.

With these rules, it is possible to construct a GRUU without requiring the maintenance of any additional state. To do that, the URI would be constructed in the following fashion:

```
user-part = "GRUU" | BASE64(E(K, (salt | " " | AOR | " " |  
instance ID)))
```

Where E(K,X) represents a suitable encryption function (such as AES with 128-bit keys) with key K applied to data block X, and the "|" operator signifies concatenation. The single space (" ") between components is used as a delimiter, so that the components can easily be extracted after decryption. Salt represents a random string that prevents a client from obtaining pairs of known plaintext and ciphertext. A good choice would be at least 128 bits of randomness in the salt.

This mechanism uses the user-part of the SIP URI to convey the encrypted AOR and instance ID. The user-part is used instead of the "opaque" URI parameter because it has the desired anonymity properties.

The benefit of this mechanism is that a server need not store additional information on mapping a GRUU to its corresponding contact. The user-part of the GRUU contains the instance ID and AOR. Assuming that the domain stores registrations in a database indexed by the AOR, the proxy processing the GRUU would look up the AOR, extract the currently registered contacts, and find the one that matches the instance ID encoded in the Request-URI. The contact whose instance ID is that instance ID is then used as the translated version of the GRUU. Encryption is needed to prevent attacks whereby the server is sent requests with fake GRUUs, causing the server to direct requests to any named URI. Even with encryption, the proxy should validate the user part after decryption. In particular, the AOR should be managed by the proxy in that domain. Should a UA send a request with a fake GRUU, the proxy would decrypt and then discard it because there would be no URI or an invalid URI inside.

While this approach has many benefits, it has the drawback of producing fairly long GRUUs.



Author's Address

Jonathan Rosenberg  
Cisco Systems  
600 Lanidex Plaza  
Parsippany, NJ 07054  
US

Phone: +1 973 952-5000

Email: [jdrosen@cisco.com](mailto:jdrosen@cisco.com)

URI: <http://www.jdrosen.net>



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