




Type:	 Bug	Status:	CLOSED
Severity:	 Major	Resolution:	Fixed
Affects Version/s:	16.15.0	Target Release	13.38.1, (6)
Component/s:	pjproject/pjsip, (1)	Version/s:	
Labels:	None	Security Level:	None
Frequency of Occurrence:	Occasional		
PJSIP Bundled:	Yes		

Description

Asterisk crashed sometimes (randomly)

Issue Links

is related to

 [ASTERISK-29219 res_pjsip_diversion: Crash if Tel URI contains History-Info](#)  CLOSED

 [SWP-11333](#) You do not have permission to view this issue

is a clone of


 [SWP-11319](#) You do not have permission to view this issue

Gerrit Reviews

No reviews matched the request. Check your Options in the drop-down menu of this sections header.

Activity

- All
- Comments
- History
- Activity
- Transitions

 Asterisk Team added a comment - 02/Dec/20 8:23 AM


Thanks for creating a report! The issue has entered the triage process. That means the issue will wait in this status until a Bug Marshal has an opportunity to review the issue. Once the issue has been reviewed you will receive comments regarding the next steps towards resolution. Please note that log messages and other files should not be sent to the Sangoma Asterisk Team unless explicitly asked for. All files should be placed on this issue in a sanitized fashion as needed.

A good first step is for you to review the [Asterisk Issue Guidelines](#) if you haven't already. The guidelines detail what is expected from an Asterisk issue report.

Then, if you are submitting a patch, please review the [Patch Contribution Process](#).

Please note that once your issue enters an open state it has been accepted. As Asterisk is an open source project there is no guarantee or timeframe on when your issue will be looked into. If you need expedient resolution you will need to find and pay a suitable developer. Asking for an update on your issue will not yield any progress on it and will not result in a response. All updates are posted to the issue when they occur.

Please note that by submitting data, code, or documentation to Sangoma through JIRA, you accept the Terms of Use present at <https://www.asterisk.org/terms-of-use/>.

 Mikhail Ivanov added a comment - 02/Dec/20 8:27 AM

crashdump

 Sean Bright added a comment - 02/Dec/20 10:37 AM - edited

Diversion header in the response has a `tel` URI which causes the crash:

```
SIP/2.0 181 Call Is Being Forwarded
Via: SIP/2.0/UDP 192.168.51.205:5060;branch=z9hG4bKJfec9fa52-fc33-4475-8f35-698aa568f3b0;received=192.168.51.205;rport=5060
From: <sip:74752429999@37.235.174.246>;tag=ebc0f9df-6d62-443d-9bfe-7b0dcc913905
To: <sip:74752718801@192.168.51.201>;tag=1801p0D659D0u8
Call-ID: d17fc6e4-278f-4613-9c4c-85840c517a0f
CSeq: 3552 INVITE
Contact: <sip:74752718801@192.168.51.201:5060>
Supported: replaces
Allow: INVITE, ACK, BYE, CANCEL, PRACK, REGISTER, INFO, REFER, NOTIFY, OPTIONS, UPDATE
Diversion: <tel:4952761626>;reason=no-answer;privacy=off
User-Agent: smg_pa_sip smg_pa_sip-3.10.1.22
Content-Length: 0
```


 Kevin Harwell added a comment - 02/Dec/20 4:26 PM

Sean Bright A most excellent find!

To easily replicate simply alter [this line](#) in the associated testsuite test to contain a `tel` URI and execute.

People

Assignee:

 Unassigned

Reporter:

 Mikhail Ivanov

Issue Participants:

Asterisk Team, Kevin Harwell, (2)

Issue Consultant:

Unassigned

Watchers:

4 Start watching this issue

Dates

Created:
02/Dec/20 8:23 AM

Updated:
13/Oct/21 6:12 AM

Resolved:
22/Dec/20 4:49 PM
