

JWT Authentication in OpenSIPS



 opensips



 VOICENTER

About me



CTO & Founders of Voicenter LTD

Voicenter provides cloud contact center solutions and services for Businesses, enterprises and telcos, With global coverage using flexible distributed topologies.

More than 15 years of business experience in establishing and managing large scale networks and information technology systems. Involved with several open source projects.



About me



Opensips Core Developer

Coding, building and integrating OpenSIPS based platforms for more than 10 years

Agenda

1. Digest Authentication limitations
2. What is JWT ?
3. AUTH_JWT OpenSIPS Module
4. JSSIP JWT implementation
5. Use cases
6. Authentication Provider Integration - OAuth
7. OAuth Demo
8. Questions

Digest Authentication limitations

1. Doing only Authentication not giving us any Authorization
2. Lack of expression
3. MD5 is not secure
4. Subscriber Table sizing problem
5. No safe way for third party auth integration
6. No ability for SSO or Dual-factor authentication.



What is JWT

JSON **W**eb **T**okens are an open, industry standard RFC 7519 method for representing claims securely between two parties (jwt.io snapshot)

Encoded PASTE A TOKEN HERE

```
eyJhbGciOiJIUzI1NiIsInR5cCI6IkpXVCJ9.eyJzdWIiOiIxMjM0NTY3ODkwIiwibmFtZSI6IkpRvY3RvZSBFdmIsIiwiaWF0IjoxNTE2MjM5MDIyfQ.t  
sNv7C2wI_Tp2xM8-CsS6kHCZ9yf-  
p3YKvtGgCWHbz
```

Decoded EDIT THE PAYLOAD AND SECRET

HEADER: ALGORITHM & TOKEN TYPE

```
{  
  "alg": "HS256",  
  "typ": "JWT"  
}
```

PAYLOAD: DATA

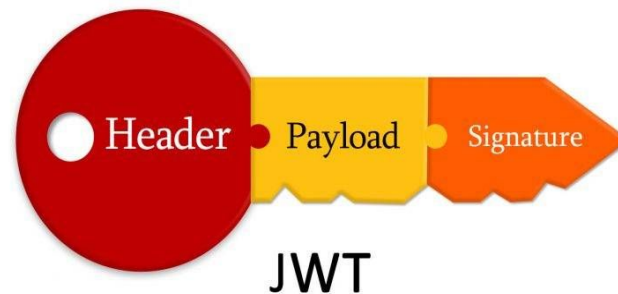
```
{  
  "sub": "1234567890",  
  "name": "Doctoe Evil",  
  "iat": 1516239022  
}
```

VERIFY SIGNATURE

```
HMACSHA256(  
  base64UrlEncode(header) + "." +  
  base64UrlEncode(payload),  
  WeLoveOpensips  
) ☐ secret ☒ base64 ☒ encoded
```

What is JWT

A well-formed JWT consists of three concatenated Base64url-encoded strings, separated by dots (.):



1. **JOSE - Header:**

contains metadata about the type of token and the cryptographic algorithms used to secure its contents.

2. **JWS - payload (set of claims):**

contains verifiable security statements, such as the identity of the user and the permissions they are allowed.

3. **JWS signature:**

used to validate that the token is trustworthy and has not been tampered with. When you use a JWT, you must check its signature before storing and using it.

JWT over SIP

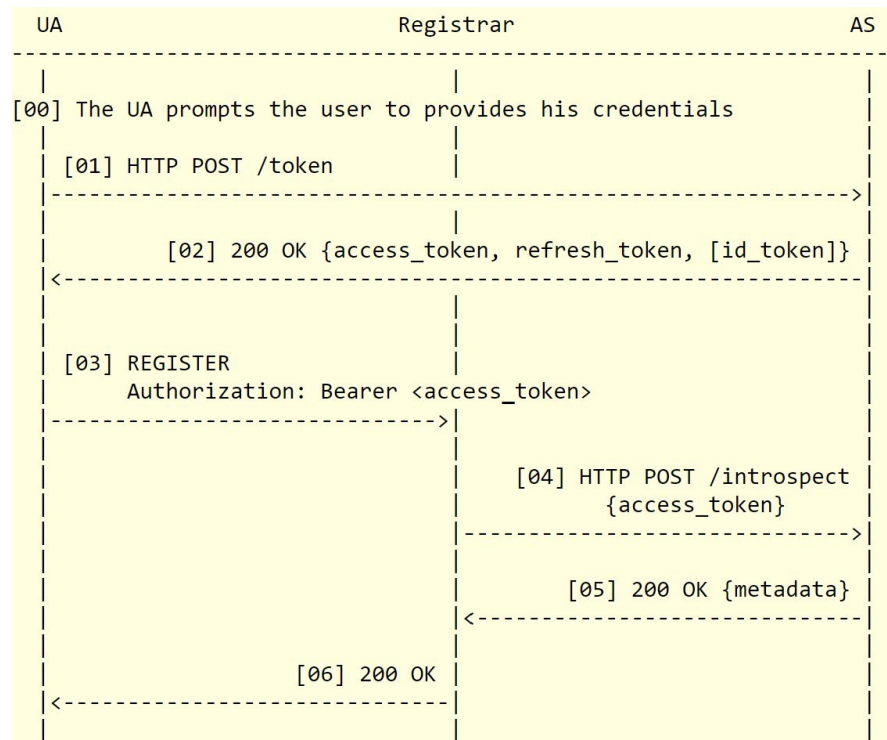
<https://tools.ietf.org/id/draft-ietf-sipcore-sip-token-authnz-02.html>



SIP Core	R. Shekh-Yusef, Ed.
Internet-Draft	Avaya
Updates: 3261 (if approved)	C. Holmberg
Intended status: Standards Track	Ericsson
Expires: January 8, 2020	V. Pascual
	webrtchacks
	July 7, 2019

Third-Party Token-based Authentication and Authorization for Session Initiation Protocol (SIP)

draft-ietf-sipcore-sip-token-authnz-02



JWT over SIP

DevTools - med...net/jssip/

Elements Console Sources Network Performance Memory Application Security Lighthouse Codeception TestTools Socket.io

Filter ☐ Hide data URLs All XHR JS CSS Img Media Font Doc WS Manifest Other ☐ Has blocked cookies ☐ Blocked Requests

Name ☐ webrt...net

Headers Messages Initiator Timing

All Enter regex, for example: (web)?socket

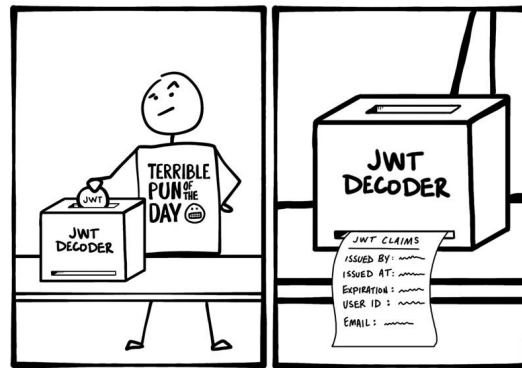
Data	Length	Time
↑ REGISTER sip: [redacted] IP/2.0 Via: SIP/2.0/WSS mttmvvo4vcre.invalid;branch=z9hG4bK4911169 Max-For...	907	23:15:17.422
↓ SIP/2.0 200 OK Via: SIP/2.0/WSS mttmvvo4vcre.invalid;received=188.120.135.182;rport=49213;branch=z9hG4bK491116...	599	23:15:17.600

```
1 REGISTER sip:c7594.ratetel.com:8888 SIP/2.0
2 Via: SIP/2.0/WSS mttmvvo4vcre.invalid;branch=z9hG4bK4911169
3 Max-Forwards: 69
4 To: <sip:[redacted]:8888>
5 From: <sip:[redacted];tag=c08ticotqd
6 Call-ID: jul1a2993tou8pdlrc57a0l
7 CSeq: 1 REGISTER
8 Authorization: Bearer eyJhbGciOiJIUzI1NiIsInR5cCI6IkpXVCJ9.eyJzdWIiOiIxMjM0NTY3ODkwIiwibmFtZSI6IkpvaG4gRG91IiwiaWF0Ijox
9 Contact: <sip:ml9igtpt@mttmvvo4vcre.invalid;transport=ws>;sip.ice;reg-id=1;sip.instance="urn:uuid:ea825621-bec2-4a3b
10 Expires: 600
11 Allow: INVITE,ACK,CANCEL,BYE,UPDATE,MESSAGE,OPTIONS,REFER,INFO,NOTIFY
12 Supported: path,gruu,outbound
13 User-Agent: JsSIP 3.3.11
14 Content-Length: 0
15
```

1 requests 0 B transferred 0 B resou

AUTH_JWT OpenSIPS modules

- `loadmodule "auth_jwt.so"`
- `modparam("auth_jwt", "db_url", "mysql://opensips:opensipsrw@127.0.0.1/opensips")`
- `modparam("auth_jwt", "profiles_table", "jwt_profiles")`
- `modparam("auth_jwt", "secrets_table", "jwt_secrets")`
- `modparam("auth_jwt", "load_credentials", "profile_info")`



AUTH_JWT OpenSIPS module - jwt_profiles table

	id INT(11)	tag VARCHAR(128)	sip_username VARCHAR(64)	profile_info VARCHAR(2048)
▶	3	space-monkey ▼	tyler.durden	{"FOO": "BAR"}

```
CREATE TABLE opensips.jwt_profiles (  
  id int(11) NOT NULL AUTO INCREMENT,  
  tag varchar(128) DEFAULT NULL,  
  sip username varchar(64) DEFAULT NULL,  
  profile info varchar(2048) DEFAULT NULL,  
  PRIMARY KEY (id)  
)
```

AUTH_JWT OpenSIPS module - jwt_secrets table

id INT(11)	corresponding_tag VARCHAR(128)	secret VARCHAR(1024)	start_ts INT(11)	end_ts INT(11)
▶ 8	space-monkey	ThefirstruleofFightclubisyoudonottalkaboutfig...	1583239007	2084325407

```
CREATE TABLE opensips.jwt_secrets(  
  id int(11) NOT NULL AUTO INCREMENT,  
  corresponding_tag varchar(128) DEFAULT NULL,  
  secret varchar(1024) DEFAULT NULL,  
  start_ts int(11) DEFAULT NULL,  
  end_ts int(11) DEFAULT NULL,  
  PRIMARY KEY (id)  
)
```

AUTH_JWT OpenSIPS module - Usage example

```
modparam("auth_jwt", "tag_claim", "iss")

if ($hdr(Authorization) =~ "Bearer") {

    $var(jwt_token) = $(hdr(Authorization){s.select,1, }{s.trim});

    if (jwt_authorize("$var(jwt_token)", $avp(decoded_jwt), $avp(out_username))) {

        if ($fU != $avp(out_username)) {

            send_reply(403, "Forbidden SIP ID");

            exit;

        }

        xlog("JWT auth - profile info = $avp(profile_info) \n");

    }

}
```

HEADER: ALGORITHM & TOKEN TYPE

```
{
  "alg": "RS256",
  "typ": "JWT"
}
```

PAYLOAD: DATA

```
{
  "iss": "space-monkey",
  "email": "space-monkey@opensips.org",
  "email_verified": true,
  "pstn_dial" : true
}
```

AUTH_JWT OpenSIPS module - jwt_authorize function

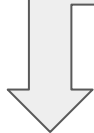
Jwt_authorize (jwt_token, out_decoded_token, out_sip_username)

The function will read the first param (jwt_token) and extract the tag claim



Try to authenticate
it against the DB
secrets

success



populates the out_decoded_token pvar with the decoded JWT, the out_sip_username with the SIP username corresponding to that JWT profile, and loads all columns listed in the load_credentials modparam.

failure

```
send_reply("401","Unauthorized");
```

JSSIP JWT implementation

🔗 **Version 3.4.0** (released in 2020-03-29)

- Add `authorization_jwt` configuration parameter (#610). Credits to @voicenter.



```
var socket = new window.JsSIP.WebSocketInterface( 'wss://webrtc.space-monkey.com.net');
```

```
var configuration = {
  sockets : [ socket ],
  session_timers: false,
  uri      : 'sip:tyler.durden@webrtc.space-monkey.net,
  authorization_jwt : "eyJhbGciOiJIUz.yyyiamnotstupid.notrealtokenxxx"
};
```

```

window.ua = new window.JsSIP.UA(configuration);
window.ua.start();

```

Use cases - How should we use it ?

- WebRTC Web systems integration
- Zero provisioning registrations
- S-S-S-O: Sip Single-Sign-On
- Advanced permission management
- Third party permission grant



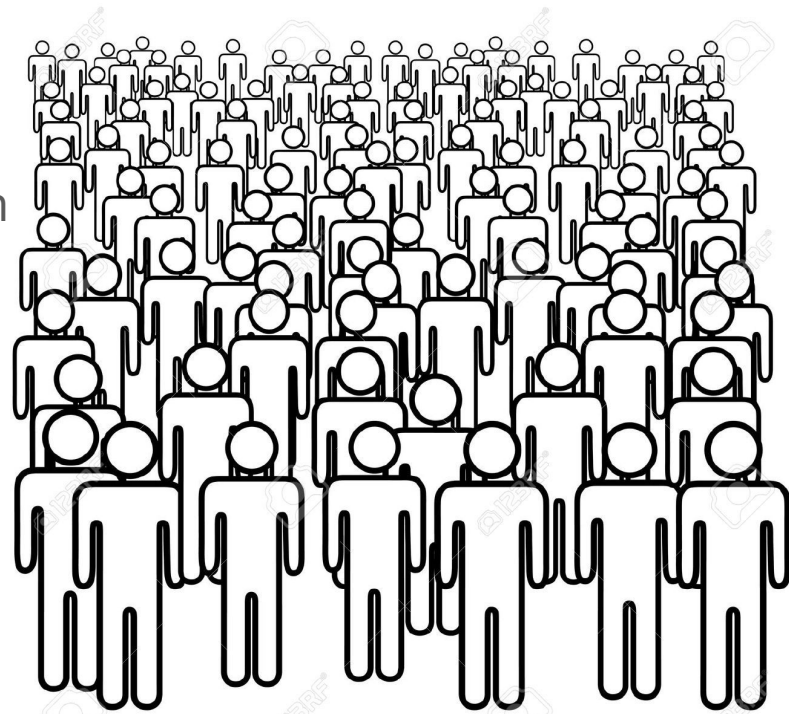
Existing application SIP integration

- CRM and ERP phone integration
- self service portal: click-to-call customers support
- Mobile app auth mechanism
- online games conferencing integration



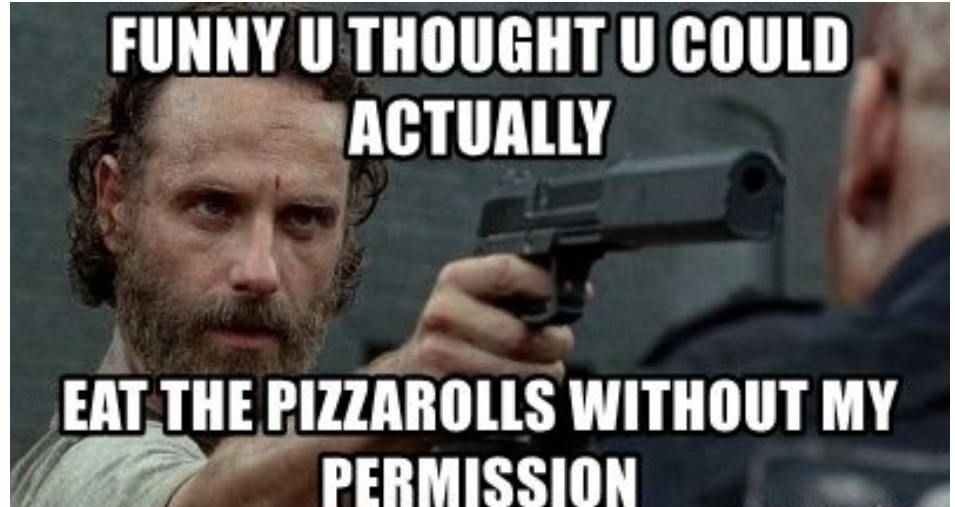
Zero provisioning registrations

1. Huge amount of potential users
2. Public access with need for authentication
3. Short time user's life cycle



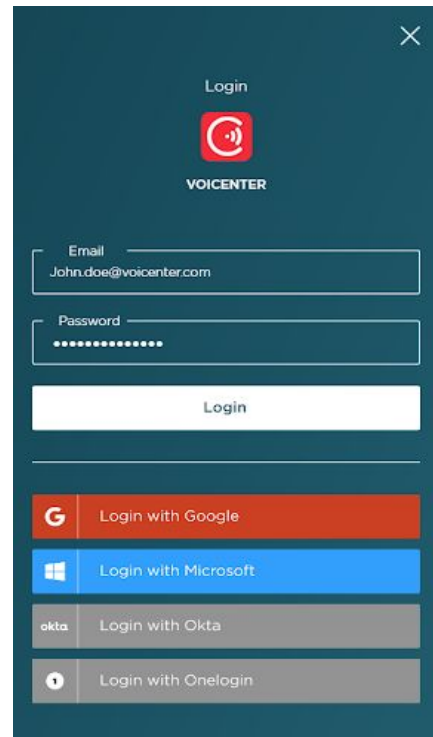
Advanced permission management

1. Call center advance call action
 - a. Eavesdropping
 - b. Whispering
 - c. Conferencing
2. International Calls restriction
3. IP based restriction
4. Conference rooms restriction



S-S-S-O - Sip Single Sign On

1. Save on login time
2. Enjoy Auth provider security policy
3. Save on IT deployment cost
4. SSO became a standard in the business application
5. Reduce human-factor related risks involved credentials handling.

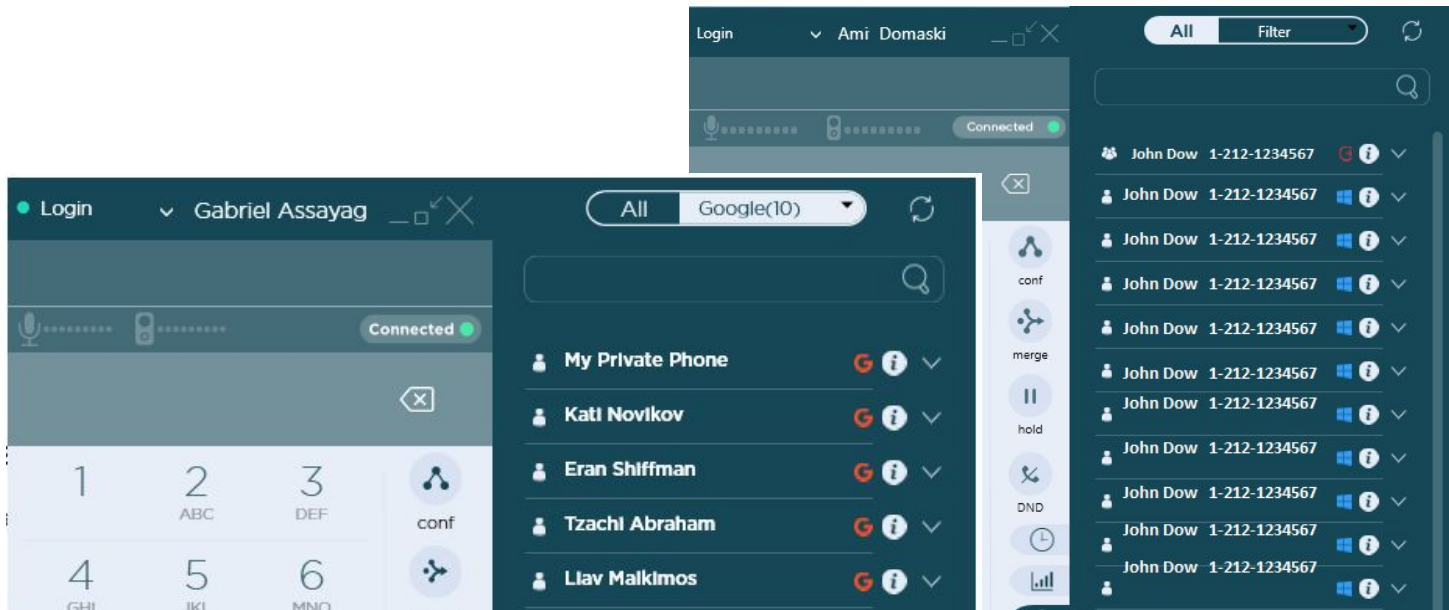
A screenshot of a web login form for Voicenter. The form has a dark teal background. At the top, it says 'Login' and features the Voicenter logo (a red circle with a white 'C' and a speech bubble) and the word 'VOICENTER'. Below this are two input fields: 'Email' with the value 'John.doe@voicenter.com' and 'Password' with masked characters. A 'Login' button is below the password field. At the bottom, there are four social login buttons: 'Login with Google' (orange), 'Login with Microsoft' (blue), 'Login with Okta' (grey), and 'Login with Onelogin' (grey).

Third party permission grant

1. Get list of contact for caller name and contact integrations
2. Store a Record on storage provider cloud on clients buckets
3. Create activity in CRM's on the call made with a client in the client activity log

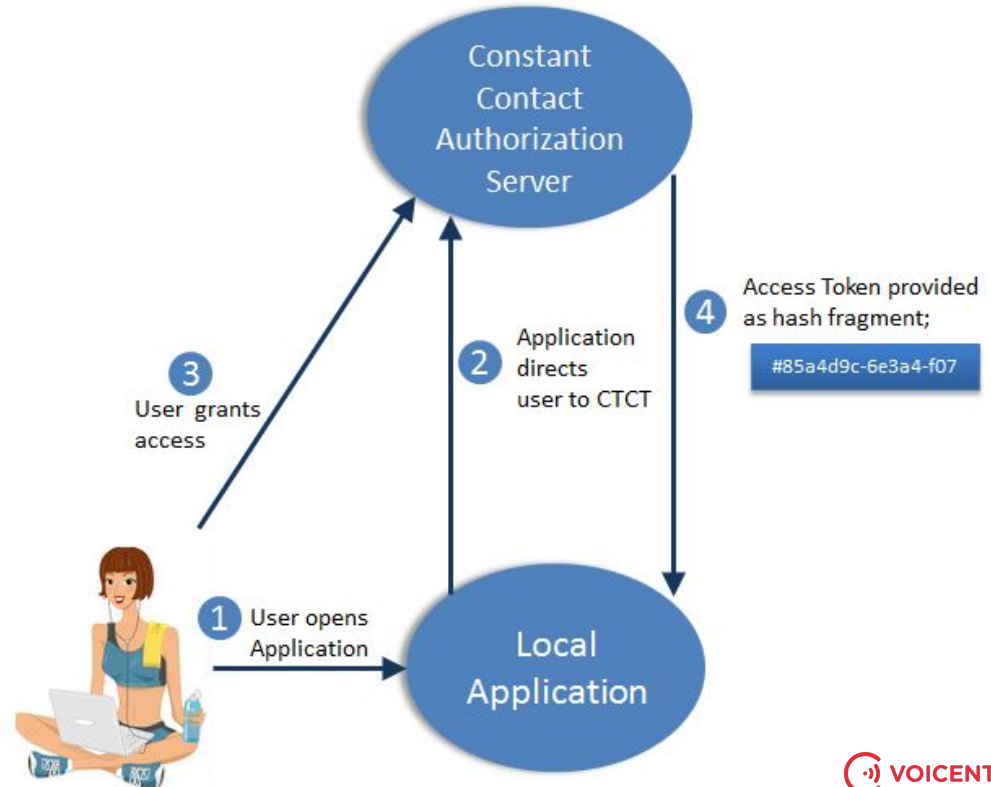


And many more...



Authentication Provider Integration - OAuth

OAuth is an open-standard authorization protocol or framework that describes how unrelated servers and services can safely allow authenticated access to their assets without actually sharing the initial, related, single logon credential.



SIP OAuth Flow

Step 1 – The user shows intent

- **Joe (user):** “Hi Voicenter. I would like to make a call and save its recording”
- **Voicenter (user application):** “Great! Let me go to Google to ask for permission.”

Step 2 – Voicenter (user application) gets permission

- **Voicenter (user application):** “Hello Google , i have a user that would like to make a call and save its recording at your storage. Can I have a request token?”
- **Google Cloud Storage (Service Provider):** “Sure. Here’s a token and a secret.”

The secret is used to prevent request forgery. The application uses the secret to sign each request so that the service provider can verify it is actually coming from the consumer application.

SIP OAuth Flow

Step 3 – The User Is Redirected to the Service Provider

- **Voicenter:** “OK, Joe. I’m sending you over to Google so you can approve. Take this token with you.”
- **Joe:** “OK!”

<Voicenter directs Joe to Google for authorization>

Step 4 – The User Gives Permission

- **Joe:** “Google, I’d like to authorize this request token that Voicenter gave me.”
- **Google:** “OK, just to be sure, you want to authorize Voicenter to make a call and store the call recording file in your Google storage account?”
- **Joe:** “Yes!”
- **Google:** “OK, you can go back to Voicenter callback ,and tell them there that they have permission to use that token ”

Google marks the request token as “good-to-go,” so when the Application requests access, it will be accepted (so long as it’s signed using their shared secret).

SIP OAuth Flow

Step 5 – The Application obtains an access token

- **Joe:** “Google gave me that token , can i register my phone ?”
- **Google:** “Sure, i can see it was sing by google , you are 200 Super OK ,”

Step 6 – The application accesses the protected resource

- **Voicenter:** “I’d like to to store this call on your storage . Here’s my JWT token!”
- **Google:** “Done!”

OAuth SIP cake recipe

- 1 Login HTML page
- 1 Phone Html page
- 3 js functions (GetConfig, Register ,Call) JSSIP implementation
- 1 Web Server backend OAuth login
- 1+ Auth provider application config
- 2 Auth provider public signature
- 1 shot of espresso



Login HTML page

/public/Login.js

```
<!DOCTYPE html>
<html lang="en">
<head>
  <meta charset="UTF-8">
  <title>Title</title>
</head>
<body>
<form action="/connect/google" method="POST" accept-charset="utf-8">
  <fieldset>
    <div>
      <button>Submit</button>
    </div>
  </fieldset>
</form>
</body>
</html>
```

Phone HTML page

/public/Phone.html

```
<!DOCTYPE html>
<html lang="en">
<head>
  <meta charset="UTF-8">
  <script src="https://jssip.net/download/releases/jssip-3.5.7.js"></script>
  <script src="call.js"></script>
  <title>Title</title>
</head>
<body>
  <button onClick="Register()"> Register</button>
  <button onClick="Call()"> Call</button>
</body>
</html>
```

3 JS functions (GetConfig,Register ,Call) JSSIP implementation

```
function GetConfig() {  
    const urlParams = new URLSearchParams(window.location.search);  
    if(urlParams.get('token'))token=urlParams.get('token');  
    console.log("token",token)  
    var configuration = {  
        sockets : [ new window.JsSIP.WebSocketInterface('wss://webrtc.officering.net:8888') ],  
        session_timers: false,  
        uri : 'sip:oauth@webrtc.officering.net',  
        authorization_jwt : "Bearer "+token  
    };  
    return configuration  
}  
  
function Register() {  
    window.ua = new window.JsSIP.UA(GetConfig());  
    window.ua.start();  
}  
  
function Call() {  
    window.ua.activeCall = window.ua.call('sip:6001@webrtc.officering.net',{mediaConstraints: {'audio': true, 'video': false}});  
    window.ua.activeCall.connection.addEventListener('addstream', (e) => {  
        console.log("addstream",e);  
        var audio = document.createElement('audio');  
        audio.srcObject = e.stream;  
        audio.play();  
    });  
}
```

/public/call.js

Web Server backend OAuth login

/server.js

```
var fastify = require('fastify')
var cookie = require('fastify-cookie')
var session = require('fastify-session')
var parser = require('fastify-formbody')
var grant = require('grant').fastify()
const path = require('path')
fastify()
  .register(cookie)
  .register(session, {secret: '01234567890123456789012345678912', cookie: {secure: false}})
  .register(parser)
  .register(grant(require('./config')))
  .register(require('fastify-static'), {root: path.join(__dirname, 'public'), prefix: '/' })
  .route({method: 'GET', path: '/', handler: async (req, res) => {
    if(req.session && req.session.grant && req.session.grant.response){
      res.redirect("/Phone.html?token="+req.session.grant.response.raw.id_token);
    }else{
      res.redirect("/Login.html");
    }
  }})
  .listen(3001)
```

Google Oauth application config

Google APIs

voicenter-tracker

Search for APIs and Services

API

Client ID for Web application

DOWNLOAD JSON

RESET SECRET

DELETE

Name *

Web Login test

The name of your OAuth 2.0 client. This name is only used to identify the client in the console and will not be shown to end users.

Client ID

7725915824[REDACTED]qvodn1cro0gld8.apps.googleusercontent.com

Client secret

9tvWAW[REDACTED]hntRH

Creation date

August 18, 2020 at 10:11:22 PM GMT+3

1

The domains of the URIs you add below will be automatically added to your [OAuth consent screen](#) as [authorized domains](#).

Authorized JavaScript origins

For use with requests from a browser

URIs

http://127.0.0.1:3000

https://summitdemo.officering.net

+ ADD URI

Authorized redirect URIs

For use with requests from a web server

URIs

http://127.0.0.1:3000/connect/google/callback

https://summitdemo.officering.net/connect/google/callback

+ ADD URI

SAVE

CANCEL

<https://console.developers.google.com/>

The Google Developers logo, featuring the multi-colored Google 'G' icon followed by the text 'Google Developers' in a sans-serif font.

12

 VOICENTER

Google Oauth application config

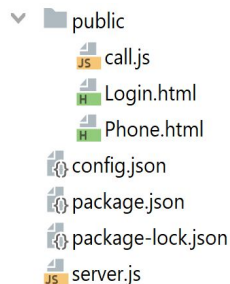
/config.js

```
{
  "defaults": {
    "origin": "http://localhost:3001",
    "transport": "session"
  },
  "google": {
    "key": "772*****d8.apps.googleusercontent.com",
    "secret": "9t*****htRH",
    "dynamic": ["scope"],
    "scope": ["openid", "email", "https://www.googleapis.com/auth/admin.directory.user.alias.readonly"],
    "response": ["raw", "jwt"],
    "callback": "/"
  }
}
```

[illegible]

Instruction checklist

- Step 1 - Make a new folder and run
 - i. `> npm init -y`
 - ii. `> npm install -s grant fastify-formbody fastify-session fastify-cookie fastify fastify-static`
- Step 2 - Create 5 files
 - i. `public/Login.html`
 - ii. `public/Phone.html`
 - iii. `public/call.js`
 - iv. `/config.js`
 - v. `/server.js`
- Step 3 - Create Google OAuth application config , and add the **key** and **secret** on `/config.js`
<https://console.developers.google.com/>
<https://developers.google.com/identity/protocols/oauth2>
- Step 4 - Import Google JWT public signature
- Step 5 - Run the server.js file :
`> node server.js`



Enjoy your cake over `http://127.0.0.1:3000`

How it will look



1>

← → ↻ ⓘ 127.0.0.1:3000/Login.html

Submit

2>

 Sign in with Google

Choose an account

to continue to [Voicenter Oauth test](#)

3>

← → ↻ ⓘ 127.0.0.1:3000/Phone.html?token=eyJhbGciOiJIUzI1NiIsInR5cGE6YWV0Ij06a

Register Call

DevTools - 127.0.0.1:3000/Phone.html?token=eyJhbGciOiJIUzI1NiIsInR5cGE6YWV0Ij06a

Elements Console Sources Network Performance Memory Application Security Lighthouse

Filter Hide data URLs All XHR JS CSS Img Media Font Doc WS Manifest Other Has blocked c

Name Headers Messages Initiator Timing

☐ webtrtc.officering.net

Data

```
1 REGISTER sip:webtrtc.officering.net SIP/2.0 Via: SIP/2.0/WSS 1r23ap1d1goa.invalid;branch=z9hG4k7
2 SIP/2.0 200 OK Via: SIP/2.0/WSS 1r23ap1d1goa.invalid;received=188.120.135.182;rport=54849;br
3 INVITE sip:6001@webtrtc.officering.net SIP/2.0 Via: SIP/2.0/WSS 1r23ap1d1goa.invalid;branch=z9hG4k7
4 REGISTER sip:webtrtc.officering.net SIP/2.0
5 Via: SIP/2.0/WSS 1r23ap1d1goa.invalid;branch=z9hG4k7051853
6 Max-Forwards: 69
7 To: <sip:oauth@webtrtc.officering.net>
8 From: <sip:oauth@webtrtc.officering.net>;tag=p8pc7o8hq7
9 Call-ID: r7rkds4odkh3oi4q0mbgc8
10 CSeq: 1 REGISTER
11 Authorization: Bearer eyJhbGciOiJIUzI1NiIsInR5cGE6YWV0Ij06a
12 Contact: <sip:2rm0conu@1r23ap1d1goa.invalid;transport=ws>;sip.ice;reg-id
13 Expires: 600
14 Allow: INVITE, ACK, CANCEL, BYE, UPDATE, MESSAGE, OPTIONS, REFER, INFO, NOTIFY
15 Supported: path, gruu, outbound
16 User-Agent: 3sSIP 3.5.7
17 Content-Length: 0
```

1 / 4 requests 0 B / 809 B transferred

Invite the media server

```
if ($json(jwt_body) != NULL && $json(jwt_body/email) != NULL) {  
    xlog("XXXX - OAuth email $json(jwt_body/email) \n");  
    append_hf("X-OAUTH: $json(jwt_body/email) \r\n");  
    $avp(email)= $json(jwt_body/email);  
    $avp(emailName)=$(avp(email){s.select,0,@});  
    $avp(emailUri)="sip:" + $avp(emailName) + "@@" + $fd;  
    uac_replace_from("$avp(emailName)","sip:$avp(emailName)@$fd");  
}
```

Call flow for ge9keo90e3kq688hmf0e (Color by Request/Response)

192.168.201.231:5060	192.168.201.27:5060	>INVITE sip:6001@webrtc.officering.net SIP/2.0
qqqqqqqqqqqqqqqqqqqqqqqq	qqqqqqqqqqqqqqqqqqqqqqqq	Record-Route: <sip:192.168.201.231;r2=on;lr;ftag=q9g1n8fkso;did=eb4.b2d4b093>
13:13:05.531497	x	INVITE (SDP) x
+0.001161	x	Record-Route: <sip:209.163.136.231:8888;transport=wss;r2=on;lr;ftag=q9g1n8fkso;did=eb4.b2d4b093>
13:13:05.532658	x	100 Trying x
+1.013622	x	Max-Forwards: 68
13:13:06.546280	x	To: <sip:6001@webrtc.officering.net>
+0.188710	x	From: shlomi <sip:shlomi@webrtc.officering.net>;tag=q9g1n8fkso
13:13:06.734990	x	Call-ID: ge9keo90e3kq688hmf0e
	x	CSSeq: 196 INVITE
	x	Contact: <sip:at7jllqp@188.120.135.182:63009;transport=ws;ob>
	x	Content-Type: application/sdp
	x	Allow: INVITE,ACK,CANCEL,BYE,UPDATE,MESSAGE,OPTIONS,REFER,INFO,NOTIFY
	x	Supported: ice,replaces,outbound
	x	User-Agent: JsSIP 3.5.7
	x	Content-Length: 1035
	x	X-OAUTH: shlomi@starkey.co.il
	x	X-APP-NAME: Conference

Thank you

