Exploring an **IMS** deployment without third-party hardware

As part of our constant effort to continually expand our knowledge base, we wanted to explore the role that an **IP multimedia Subsystem (IMS)** plays on a communications network.

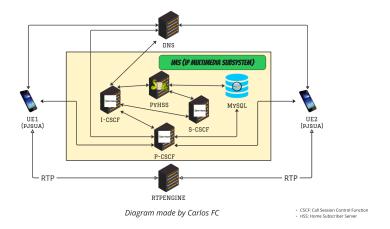
However we quickly realize that most educational literature assume a fiscal setup with actual hardware and equipment. Is it not possible, and even preferable, to get a initial understanding and minimal interactions with this technologies before diving into expensive purchasing?

We resolved ourselves to do precisely that by trying to get a call working, on a single host setup, making use of only software-based solutions.

But what is an IMS?

IMS is an IP-based network subsystem developed by 3GPP to provide multimedia services, such as voice, video, messaging, and conferencing, over both mobile and fixed-access networks. IMS enables interoperability between different technologies and facilitates the transition from traditional circuit-switched (CS) networks to fully packet-based (IP) networks.

Key components



• Call Session Control Function (CSCF)

Call control and session management. Composed of:

- Proxy-CSCF (P-CSCF): Entry point to the IMS network.
- Interrogating-CSCF (I-CSCF): Routes SIP request towards the S-CSCF and assigns it to a user performing registration.
- Serving-CSCF (S-CSCF): Session control services, including acting as a registrar.

• Home Subscriber Server (HSS)

Database containing user and subscription related information.

Protocols

Session Initiation Protocol (SIP)

An application-layer control (signaling) protocol for creating, modifying, and terminating sessions with one or more participants. These sessions include Internet telephone calls, multimedia distribution, and multimedia conferences.¹

Digest Using Authentication and Key Agreement (AKA)

AKA is a challenge-response based mechanism that uses symmetric cryptography to authenticate **User Equipments (UEs)**. When working with **IMS** the appropriate credentials are stored both on the server itself and on the user device.²

Diameter Protocol

The Diameter base protocol is intended to provide an Authentication, Authorization, and Accounting (AAA) framework for applications such as network access or IP mobility in both local and roaming situations.³

The **Diameter SIP application** allows a client to request authentication and authorization information for **SIP-based IP multimedia services**. Furthermore, it provides extra functions, such as the ability to download or receive updated user profiles, or rudimentary routing functions that can assist a **SIP server** in finding another one allocated to the user.⁴

Making a call

On a 4G/5G network a call between two UE requires some setup and can be split into three different stages:

¹Rosenberg, J., Schulzrinne, H., Camarillo, G., Johnston, A., Peterson, J., Sparks, R., Handley, M., and E. Schooler, "SIP: Session Initiation Protocol", RFC 3261, DOI 10.17487/RFC3261, June 2002, https://www.rfc-editor.org/info/rfc3261.

²Niemi, A., Arkko, J., and V. Torvinen, "Hypertext Transfer Protocol (HTTP) Digest Authentication Using Authentication and Key Agreement (AKA)", RFC 3310, DOI 10.17487/RFC3310, September 2002, https://www.rfc-editor.org/info/rfc3310.

³Fajardo, V., Ed., Arkko, J., Loughney, J., and G. Zorn, Ed., "Diameter Base Protocol", RFC 6733, DOI 10.17487/RFC6733, October 2012, https://www.rfc-editor.org/info/rfc6733.

⁴Garcia-Martin, M., Ed., Belinchon, M., Pallares-Lopez, M., Canales-Valenzuela, C., and K. Tammi, "Diameter Session Initiation Protocol (SIP) Application", RFC 4740, DOI 10.17487/RFC4740, November 2006, https://www.rfc-editor.org/info/rfc4740.

1. Attach process

Refers to the initial authentication and IP assignment of a **UE**. This process occurs primarily outside of the **IMS**.

2. Initial registration

The issuing of a REGISTER request (SIP) by a connecting UE, and the authorization process involving AKA and Diameter.

3. Call setup

A normal SIP call setup routed by the IMS. Usually an RTP proxy is used to take care of media communication between the devices

IMS core setup

This approach consist of a bare bones \mathbf{IMS} deployment, making use of the following services:

- OpenSIPS IMS CE
- rtpengine
- PyHSS

Here, we discard the **Core Network** in which the **IMS** is supposed to run and generate **SIP trafic** using 2 **PJSUA** clients that simulate our endpoint devices.

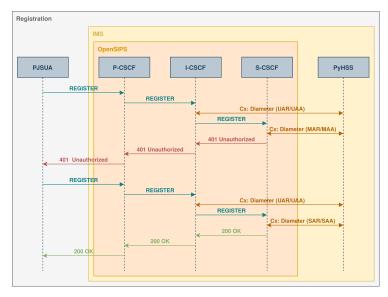
In order to correctly authenticate some arguments are needed:

```
pjsua \
--registrar=sip:ims.mnc001.mcc001.3gppnetwork.org \
--id=sip:<imsi>@ims.mnc001.mcc001.3gppnetwork.org \
--nameserver=<dns> \
--proxy sip:pcscf.ims.mnc001.mcc001.3gppnetwork.org \
--realm=ims.mnc001.mcc001.3gppnetwork.org \
--username=<imsi>@ims.mnc001.mcc001.3gppnetwork.org \
--password="$(echo <aka_k> | xxd -r -p)" \
--aka-op="<aka_op>" \
--aka-amf="<aka_amf>" \
--use-ims
```

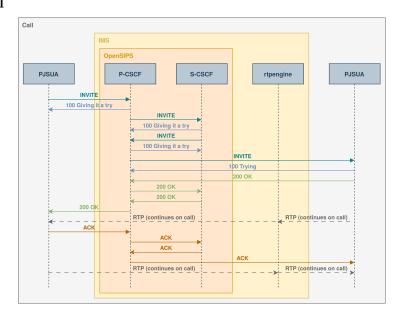
6Note

Keep in mind that --aka-op and --aka-amf options weren't part of PJSUA cli interface until this commit and are only available for version 2.16 and onward.

Registration



Call



Simulate a 4G network

We can also try to more fitfully recreate a real deployment by using the **docker open5gs** repo as a starting point. One of the included setup describes a 4G deployment to which a simulated **UE** can connect to. However, as is there is no way to generate **SIP** traffic from that device to the **IMS**, so to fix that we'll need to use the **srslte** image as a base on top of which we'll

build our \mathbf{PJSUA} clients and route their traffic trough the \mathbf{TUN} interface that encapsulates and simulates the radial network.

Conclusion

We have demonstrated that for simple setups focusing on the core functionalities of \mathbf{IMS} it is totally possible to use simulated devices and local setups without extra hardware, there's also promise in the simulation of radial networks with tools like \mathbf{ZeroMQ} and \mathbf{srsRAN} for more complete and in depth setups.