Master of Science in Internetworking, University of Alberta

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Introduction to Elastix Server

Elastix is an Open Source Software to establish Unified Communications. Elastix goal is to incorporate all the communication alternatives, available at enterprise level into a unique solution. Elastix not only provides telephony, it integrates communication alternatives to make an organization's environment more productive and efficient. From the last century Telephony was the traditional way that leads communications, that's why many users are focusing on their requirements to establish telephony communications in their organizations.

Some of the basic Features of Elastix include:

- Voicemail
- Fax-to-email
- Support for soft phones
- Web Interface Configuration
- Virtual conference rooms
- Call recording
- Least Cost Routing
- Extension Roaming
- PBX Interconnection
- Caller ID
- CRM
- Advance Reports

Elastix is an open source entrepreneur tool that can be used for commercial or personal purposes subject to the conditions as described under it's license GPLv2.All the Elastix software are full version and free to be used by anyone under it's license.

Elastix include the following communication media:

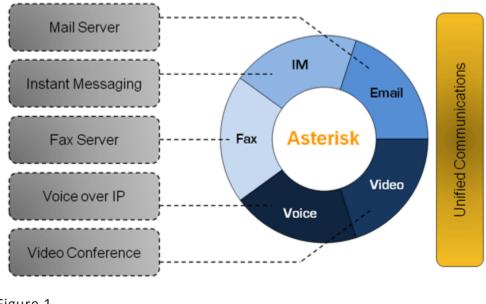


Figure 1 Ref: -<u>http://Elastix.org/</u>

Introduction to High Availability clustering with DRBD + Heartbeat

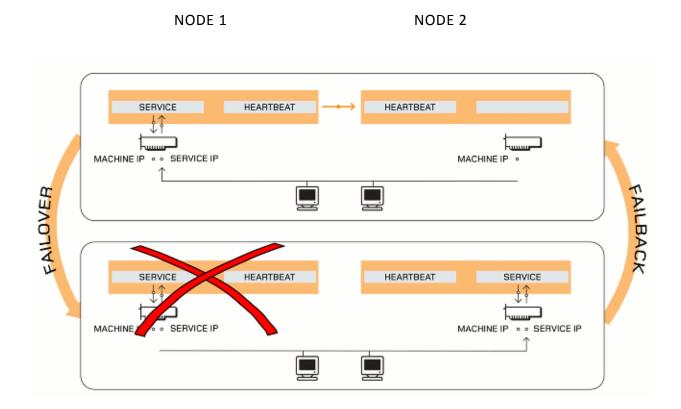
High availability clustering is a system design principle for increased availability. One of the most common implementations for HA is redundancy with failover. For example, we can define multiple paths to a given resource. High availability clustering is used with Heartbeat and Distributed Replicated Block Device (DRBD).

Introduction to DRBD and Heartbeat

The Distributed Replicated Block Device (DRBD) provides a networked version of data mirroring, classified under the redundant array of independent disks (RAID). DRBD effectively puts mirrored file system beneath the database. DRBD helps all the slave machines in replicating the changes which take place at any point of time on master

server whose ip addresses are defined in <u>drbd.conf</u>file. We can define as many machines as we want as a backup server by giving them priority.

Heartbeat is a daemon which runs on all the machines in the high availability cluster. Heartbeat communicates with all the nodes which are participating in high availability cluster and decides the role of the nodes.





The upper block in this image shows that the node 1 in the left hand side is active and the node 2 in the right hand side is passive. All the machines are getting their ip addresses from node 1 and there is heartbeat running between both of the nodes.

In the next block, as heartbeat stops between two nodes, node 2 is now acting as an active node and the machine in the left hand side is acting as standby. All the machines are getting their ip addresses from node 2. Services like ip addressing, mysql and http can be moved between the nodes without any change. Heartbeat is like a pulse between the nodes taking part in high availability clustering, which checks the status of the active server in regular intervals of time and when the heartbeat stops, either by system failure or because of administrative action there will be a failover/failback condition.

DRBD gives a flexibility of synchronizing data between the nodes from fully synchronous to asynchronous. There are three different replication modes supported by Distributed Replicated Block Device.

- **Protocol A**: It is an asynchronous replication protocol which is used for long distance replication scenarios. When this protocol is used with the combination of DRBD proxy it makes an effective recovery solution. As the local write operation is finished the replication packet is placed in TCP send buffer, but whenever there is forced failover then the data loss may occur. The data on the secondary servers is consistent. However the most recent changes before the crash can be lost.
- **Protocol B**: It is a semi-synchronous replication protocol used in DRBD. In this case there is no data loss encountered during forced failover, but if there occurs continues power failure on both the nodes and irreversible destruction of the primary nodes data happens then there will be data loss.
- **Protocol C**: It is a fully synchronous replication protocol. Data loss is inevitable even with this replication protocol if both nodes are irreversibly destroyed at the same time. This is because the local write operation is considered to be completed only when the write operation is confirmed on both the active and the passive nodes. Protocol C is the most commonly used protocol in DRBD because of its reliability.

Note: In this scenario we are using Protocol A.

Failover Solution For Elastix Servers over Wide Area Network

We will start with step by step installation of Elastix Servers.

Note: HARD DRIVES ON BOTH SERVERS MUST HAVE EXACTLY THE SAME CAPACITY.

- 1. Boot from Elastix 2.0 latest stable version cd.
- 2. Choose default language for installation and for keyboard language.
- 3. Choose: Create custom Layout, for hard disk partition and erase all previous partitions.
- 4. Create: swap partition for 2048 MB size
 /boot for 100 MB with file system ext3
 / For 150 GB with file system ext3
- 5. Configure network interface and also configure manually the hostname for both the nodes.
 For Master: master.example.com (on both the servers)
 For Slave: slave.example.com
- 6. Then select the time zone and the root password for both the servers.
- After the installation process server will reboot and ask for the root password for Mysql and for admin user which can be used for accessing the Elastix web interface of servers.
- 8. Execute from the shell: yum update -y Elastix (on both the servers) yum update -y
- Note: this will update Elastix server to the latest stable version and add-ons.

Now we have to create the partition that will contain replication data on both the servers.

9. Fdisk /dev/sdb (in our scenario we are using second hard disk)
 Create a new partition (n)

Primary partition (p) (on both the servers) Partition number (4) already selected by default Press (w) this will write the partition and will exit.

10. Then reboot both the servers.

}

11. Define the file system for the new partition we have just created: sdb4

Command: mkfs.ext3 /dev/sdb4 (on both servers)

- 12. Now for synchronization typeCommand: dd if=/dev/zero bs=1M count=1 of=/dev/sdb4; sync(On both the servers)
- 13. We need to install the packages of drbd, heartbeat, openhpi
- Command: yum install drbd83 kmod-drbd83 heartbeat openhpi -y (On both the servers)
- 14. In this step we have to define the nodes on both servers and assign the ip addresses to the master and slave servers
 Edit /etc/hosts
 172.16.2.2 master.example.com
 172.16.2.5 slave.example.com
- 15. Now we have to define the protocol and the locations of the servers which we want to be synchronized with each other. Edit /etc/drbd.conf

```
resource "r0" {
    protocol C;
    disk{on-io-error pass_on ;}
    startup{wfc-timeout 5; degr-wfc-timeout 3 ;}
    syncer{rate100M ;}
  on master.example.com {
        device /dev/drbd0;
    disk /dev/sdb4;
    address172.16.2.2:7789;
    meta-disk internal;
    on slave.example.com {
        device /dev/drbd0;
    }
}
```

disk /dev/sdb4; address 172.16.2.5:7789; meta-disk internal; }}

- 16. Here we have to execute the following command on both the servers drbdadm create-md r0 Then reboot both of the nodes.
- 17. Now execute

service drbd start (on both servers) For starting synchronization on both the servers, initially both the servers will be secondary and when we execute <u>service drbd status</u>. It will give:-Secondary/Secondary (on both servers)

- Execute drbdsetup /dev/drbd0 primary -o (on master server only) For making one of the two servers a primary node.
- 19. We can determine the role of a server by executing the following command: drbdadm role r0

The primary server should return: Primary/Secondary The secondary server should return: Secondary/Primary

- 20. We can mount the virtual partition /dev/drbd0, but first we must format the partition with ext3 format by using the following commands. mke2fs –j /dev/drbd0 mkdir /replica mount /dev/drbd0 /replica (only on master server)
- 21. Now we will make a directory named as replica only on secondary server mkdir /replica
- 22. Install Elastix call center module on both servers. yum install Elastix-callcenter -y

23. Only on Master server execute these commands cd /replica

tar -zcvf etc-asterisk.tgz /etc/asterisk tar -zcvf var-lib-asterisk.tgz /var/lib/asterisk tar -zcvf var-www.tgz /var/www/ tar -zcvf var-lib-mysql.tgz /var/lib/mysql/ tar -zcvf opt-Elastix-dialer.tgz /opt/Elastix/dialer/ tar -zcvf usr-lib-asterisk.tgz /usr/lib/asterisk/

tar -zxvf etc-asterisk.tgz tar -zxvf var-lib-asterisk.tgz tar -zxvf var-www.tgz tar -zxvf var-lib-mysql.tgz tar -zxvf opt-Elastix-dialer.tgz tar -zxvf usr-lib-asterisk.tgz

rm -rf etc-asterisk.tgz rm -rf var-lib-asterisk.tgz rm -rf var-www.tgz rm -rf var-lib-mysql.tgz rm -rf opt-Elastix-dialer.tgz rm -rf usr-lib-asterisk.tgz

rm -rf /etc/asterisk rm -rf /var/lib/asterisk rm -rf /var/www/ rm -rf /var/lib/mysql/ rm -rf /opt/Elastix/dialer/ rm -rf /usr/lib/asterisk/

In -s /replica/etc/asterisk/ /etc/asterisk In -s /replica/var/lib/asterisk/ /var/lib/asterisk In -s /replica/var/www/ /var/www In -s /replica/var/lib/mysql/ /var/lib/mysql In -s /replica/opt/Elastix/dialer/ /opt/Elastix/dialer In -s /replica/usr/lib/asterisk/ /usr/lib/asterisk

- 24. Only on slave server execute the following commands rm -rf /etc/asterisk
 - rm -rf /var/lib/asterisk rm -rf /var/lib/mysql/ rm -rf /var/lib/mysql/ rm -rf /opt/Elastix/dialer/ rm -rf /usr/lib/asterisk/ In -s /replica/etc/asterisk/ /etc/asterisk In -s /replica/etc/asterisk/ /var/lib/asterisk In -s /replica/var/lib/asterisk/ /var/lib/asterisk In -s /replica/var/lib/mysql/ /var/lib/mysql In -s /replica/opt/Elastix/dialer/ /opt/Elastix/dialer In -s /replica/usr/lib/asterisk/ /usr/lib/asterisk
- 25. In this step stop all the services that heartbeat can control. (on both the servers)

For that execute the following:

service asterisk stop service mysqld stop service httpd stop service Elastixdialer stop service dahdi stop chkconfig asterisk off chkconfig mysqld off chkconfig httpd off chkconfig Elastixdialer off chkconfig dahdi off

Note: From here we are going to configure high availability clustering between the nodes.

There are three important files we need to take in consideration while configuring high availability clustering.

a. ha.cf b. haresources

c. authkeys

26. First of all we have to create /etc/ha.d/ha.cf (on both the servers)

debugfile /var/log/ha-debug logfile /var/log/ha-log logfacility local0 keepalive 2 deadtime 20 warntime 10 initdead 40 udpport 694 bcast eth0 # Linux auto_failback on node master.example.com

Warn time issues a warning that a node is no longer available.

Dead time is the time after which Heartbeat considers a node confirmed dead.

Initdead is the maximum time a master server waits for other nodes to check in at cluster startup.

Keepalive sets the interval at which Heartbeat keep-alive packets are sent to other nodes.

The node option identifies cluster members.

27. Create /etc/ha.d/haresources (on both servers)

master.example.com drbddisk::r0 Filesystem::/dev/drbd0::/replica::ext3 IPaddr::192.168.1.3/24/eth0/172.16.2.10 dahdi asterisk mysqld httpd Elastixdialer

(This is a single line), 172.16.2.10 will be the floating IP address.

All the services which we want to start are defined in this file. This file should be same on each and every node in the cluster.

28. Create /etc/ha.d/authkeys (on both servers) auth 1 1 sha1 Elastix

We can use sha1 or md5 whichever we want for security and Elastix is the actual authentication key.

Till here our entire important file required for high availability clustering has been created on all of the nodes participating in the cluster.

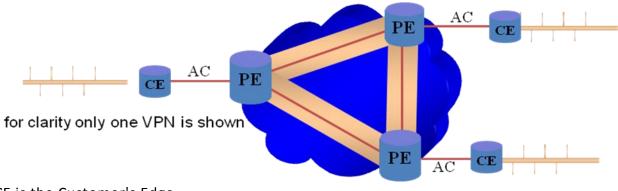
29. Execute the following commands:

5	
chmod 600 /etc/ha.d/authkeys	(both servers)
chkconfig heartbeat on	(both servers)
cd /	(only on the primary server)
umount /replica	(only on the primary server)
service drbd restart	(both servers)
drbdadm role r0	(both servers)
Service heartbeat restart	(both servers)

This configuration can work on local area network perfectly fine, where both the servers and phones are at the same location. But there is an issue with Elastix server that they can't work on different networks. We performed two different tests to check the behavior of the servers over Wide area network which are explained below.

Test 1

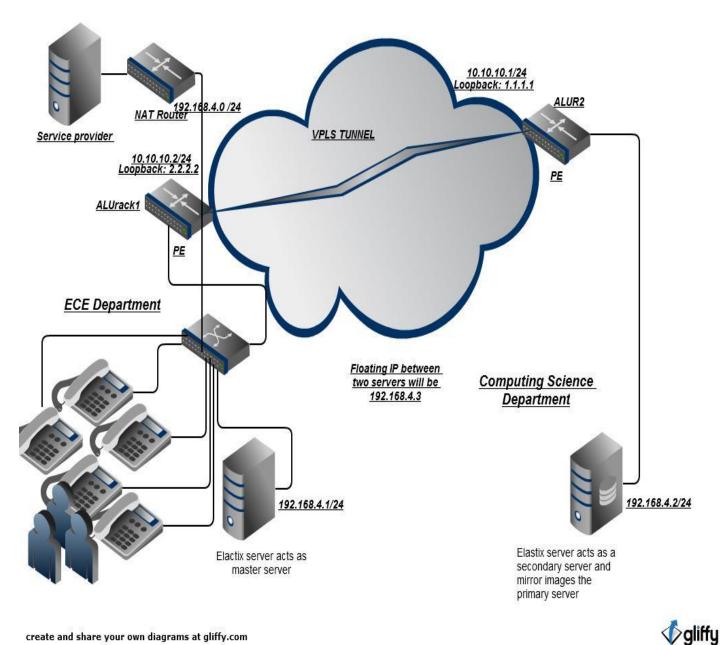
Elastix servers can only synchronize with each other within the cluster if, all the nodes are on the same network. Keeping this in mind we used VPLS (Virtual Private LAN Service). VPLS emulates a Local Area Network over an MPLS network, by connecting geographically dispersed sites through pseudo-wires. VPLS helps to share the same Ethernet broadcast domain between different sites.



CE is the Customer's Edge PE is the Service Providers Edge

There will be a MPLS tunnel between each pair of PE's which is full mesh. In our test we have used two of the Alcatel-Lucent routers as PE's.

Below given is the network diagram we have used for VPLS connectivity between two servers.



create and share your own diagrams at gliffy.com

Figure 3

Configuration for PE1

#----echo "System Configuration" #----system name "PE1" ccm 1 exit #----echo "Card Configuration" #----card 1 card-type iom-9g mda 1 mda-type c1-1gb-sfp exit mda 2 mda-type c8-10/100eth-tx exit mda 4 mda-type c2-oc12/3-sfp exit exit

#----echo "Port Configuration" #----port 1/2/1 ethernet mtu 1536 exit no shutdown exit port 1/2/2 ethernet mode access exit no shutdown exit port 1/2/3 ethernet exit no shutdown exit #----echo "Router (Network Side) Configuration" #----router interface "system"

address 2.2.2.2/32
exit
interface "toALUR2"
address 10.10.10.2/24
port 1/2/1
exit
#
echo "OSPFv2 Configuration"
#
ospf 2.2.2.2
area 0.0.0.0
interface "system"
exit
interface "toALUR2"
exit
exit
exit
#
echo "LDP Configuration"
#
ldp
interface-parameters
interface "toALUR2"
exit
exit

targeted-session
exit
exit
exit
#
echo "Service Configuration"
#
service
customer 1 create
description "Default customer"
exit
sdp 12 mpls create
far-end 1.1.1.1
ldp
keep-alive
no shutdown
exit
no shutdown
exit
vpls 5 customer 1 create
stp
no shutdown
exit
sap 1/2/2 create

exit
mesh-sdp 12:5 create
exit
no shutdown
exit
exit
#
echo "Router (Service Side) Configuration"
#
router
#
echo "OSPFv2 Configuration"
#
ospf 2.2.2.2
exit
exit
exit all
Finished SUN FEB 05 21:16:04 2012 UTC

Configuration for PE2

#
echo "System Configuration"
#

system
name "PE2"
ccm 1
exit
#
echo "Card Configuration"
#
card 1
card-type iom-9g
mda 1
mda-type c1-1gb-sfp
exit
mda 2
mda-type c8-10/100eth-tx
exit
mda 4
mda-type c2-oc12/3-sfp
exit
exit
#
echo "Port Configuration"
#
port 1/2/1
ethernet
mtu 1536

exit
no shutdown
exit
port 1/2/2
ethernet
mode access
exit
no shutdown
exit
port 1/2/3
ethernet
exit
no shutdown
exit
exit
#
echo "Router (Network Side) Configuration"
#
router
interface "system"
address 1.1.1.1/32
exit
interface "toALUrack1"
address 10.10.1/24
port 1/2/1

exit
#
echo "OSPFv2 Configuration"
#
ospf 1.1.1.1
area 0.0.0.0
interface "system"
exit
interface "toALUrack1"
exit
exit
exit
#
echo "LDP Configuration"
echo "LDP Configuration"
echo "LDP Configuration" #
echo "LDP Configuration" #
echo "LDP Configuration" # ldp interface-parameters
echo "LDP Configuration" # ldp interface-parameters interface "toALUrack1"
echo "LDP Configuration" # ldp interface-parameters interface "toALUrack1" exit
echo "LDP Configuration" # ldp interface-parameters interface "toALUrack1" exit exit
echo "LDP Configuration" # Idp interface-parameters interface "toALUrack1" exit exit targeted-session

#----echo "Service Configuration" #----service customer 1 create description "VoIP" exit sdp 12 mpls create far-end 2.2.2.2 ldp keep-alive no shutdown exit no shutdown exit vpls 5 customer 1 create stp no shutdown exit sap 1/2/2 create exit mesh-sdp 12:5 create exit no shutdown exit

#-----

echo "OSPFv2 Configuration"

#-----

ospf 1.1.1.1

exit

exit

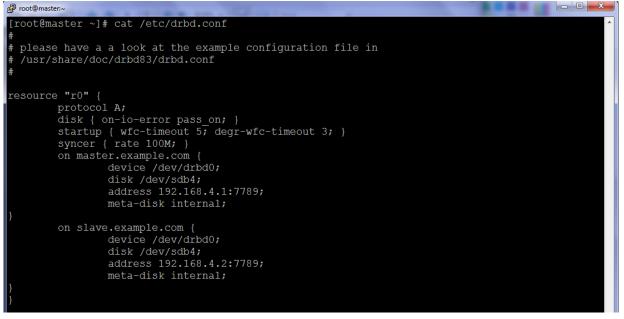
exit

Now there is VPLS tunnel running perfectly fine between Router PE1 and PE2.

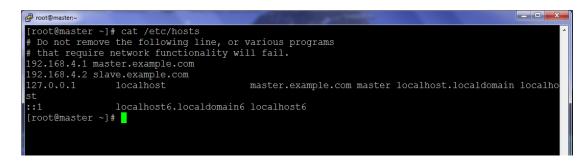
DRBD configuration on master/slave server

This file defines the location of the memory to be synchronized and the ip address of the master and slave nodes.

Note: This should be same on all the nodes within the cluster.



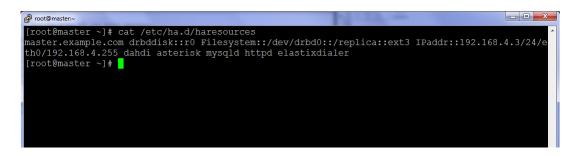
Here we are defining the host names of all the nodes in the cluster, which can communicate with each other. It should be same on all the nodes within the cluster.



These are the three most important files for the formation of high availability cluster.

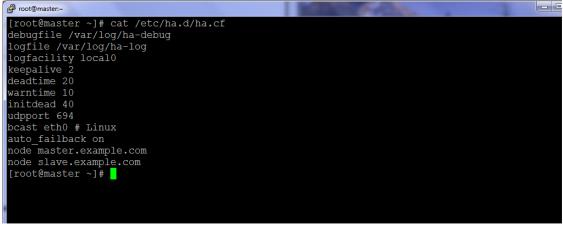
1. /etc/ha.d/haresources is a file where we have to define all the services which we want to run within the cluster. Floating ip address is also defined here.

NOTE: Everything written in this file should be in single line and should be the same on all the nodes otherwise it will not work.



 As explained earlier in this file we will define keep alive, dead time, warn time, on the nodes which are participating. We can also declare auto failback condition as per our

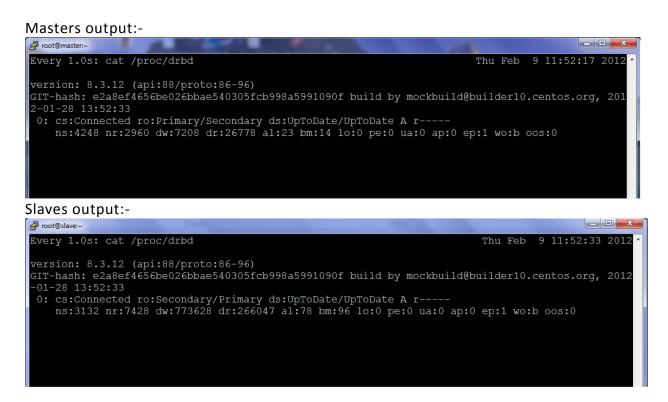
requirements.



Status and behavior of servers

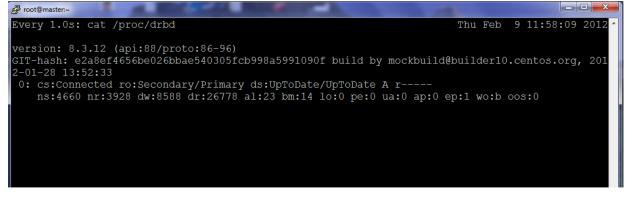
Now it's time to check the synchronization status between two servers:

On Master/Slave servers: watch -n 1 cat /proc/drbd

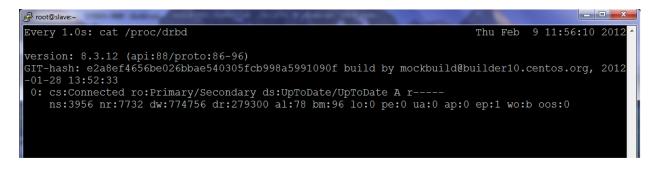


This shows that both the servers are connected and are up to date.

Command: service heartbeat stop (on master server, manually stopping the heartbeat)



When heart beat stops on master server our master server becomes secondary server and the slave server starts acting like a primary server as shown below and all the phones get registered with the secondary server with the help of floating ip address.



In our first test everything is working fine, both the servers are fully synchronized with the use of VPLS tunneling, but VPLS comes with cost. This is a service provided by the service providers, that's why we used Layer 2 Tunneling protocol v3 (L2TPv3) in our next test.

TEST 2

We are going to use Layer 2 Tunneling protocol v3 (L2TPv3) for the communication between two servers in this test. L2TPv3 provides high-speed Layer 2 tunneling or VPN services to end-user. L2TPv3 tunnels are available with the IOS basic IP package. It has combined the best features of two existing tunneling protocols: Cisco's Layer 2 Forwarding (L2F) and Microsoft's Point-to-Point Tunneling Protocol (PPTP). L2TPv3 corresponds to the data-link layer and use frames as its unit of exchange. L2TP operate on layer 2 and encapsulates the payload in a PPP frame to be sent across an internetwork.

Equipment used in Test 2

- 3 Cisco 2600 Routers
- 2 Cisco 2900 Routers
- 2 Cisco Catalyst 3750G switches
- 2 Sun Blade servers with Elastix 2.0 installed on them
- 2 IP Phones

Network Diagram for test 2.

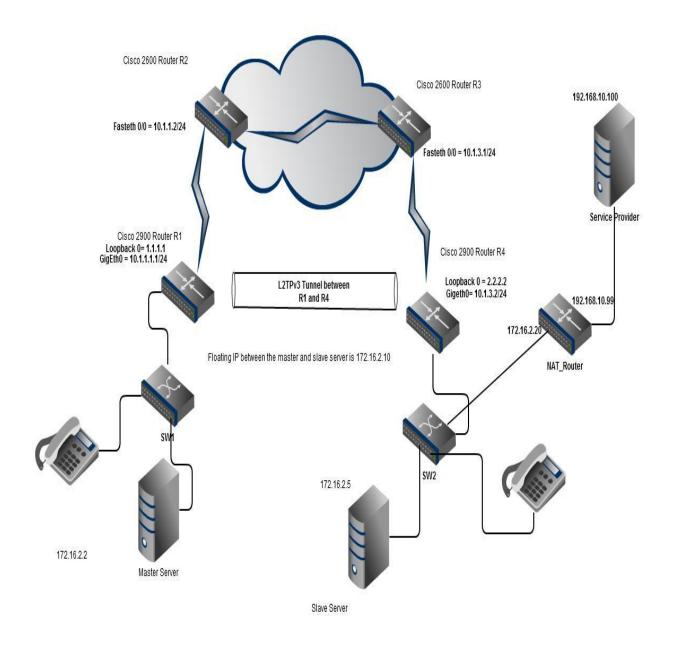


Figure 4

Configuration for Router R1

```
hostname R1
!
I
pseudowire-class MYPW
encapsulation l2tpv3
ip local interface Loopback0
L
I
ļ
interface Loopback0
ip address 1.1.1.1 255.255.255.255
ļ
!
interface GigabitEthernet0/0
ip address 10.1.1.1 255.255.255.0
duplex auto
speed auto
!
!
interface GigabitEthernet0/1
no ip address
duplex auto
speed auto
nokeepalive
xconnect 2.2.2.2 1 pw-class MYPW
!
!
interface GigabitEthernet0/2
noip address
shutdown
duplex auto
speed auto
!
!
routereigrp 100
network 0.0.0.0
ļ
```

ip forward-protocol nd

Configuration for Router R2

```
hostname R2
!
I
interface FastEthernet0/0
ip address 10.1.1.2 255.255.255.0
duplex auto
speed auto
!
interface Serial0/0
noip address
!
interface Serial0/1
ip address 10.1.2.1 255.255.255.0
clock rate 8000000
!
routereigrp 100
network 0.0.0.0
no auto-summary
!
ip http server
ip classless
I
I
voice-port 1/0/0
l
voice-port 1/0/1
I
ļ
line con 0
line aux 0
linevty 04
!
L
end
```

Configuration for Router R3

```
hostname R3
l
Ţ
l
interface FastEthernet0/0
ip address 10.1.3.1 255.255.255.0
duplex auto
speed auto
interface Serial0/0
ip address 10.1.2.2 255.255.255.0
interface FastEthernet0/1
noip address
shutdown
duplex auto
speed auto
!
interface Serial0/1
noip address
encapsulationppp
!
routereigrp 100
network 0.0.0.0
no auto-summary
!
voice-port 1/0/0
l
voice-port 1/0/1
ļ
!
line con 0
line aux 0
linevty 04
!
end
```

Configuration for Router R4

```
hostname R4
I
pseudowire-class MYPW
encapsulation l2tpv3
ip local interface Loopback0
I
interface Loopback0
ip address 2.2.2.2 255.255.255.255
ļ
interface GigabitEthernet0/0
ip address 10.1.3.2 255.255.255.0
duplex auto
speed auto
!
!
interface GigabitEthernet0/1
noip address
duplex auto
speed auto
nokeepalive
xconnect 1.1.1.1 1 pw-class MYPW
!
!
interface GigabitEthernet0/2
noip address
shutdown
duplex auto
speed auto
1
routereigrp 100
network 0.0.0.0
!
ip forward-protocol nd
Т
```

L2TP is configured between Router R1 and Router R4, as shown in the network diagram.

Configuration for Switch SW1

```
hostname SW1
I
l
interface GigabitEthernet1/0/1
switchport mode access
L
interface GigabitEthernet1/0/2
switchport mode access
I
interface GigabitEthernet1/0/3
switchport access vlan 10
!
interface GigabitEthernet1/0/6
switchport access vlan 10
switchport mode access
ļ
interface GigabitEthernet1/0/23
speed 100
duplex full
!
ļ
interface Vlan10
ip address 172.16.2.4 255.255.255.0
noip route-cache
standby 10 ip 172.16.2.1
standby 10 timers 2 4
standby 10 preempt
!
ip default-gateway 192.168.1.1
ip classless
ip http server
l
l
l
!
control-plane
!
L
```

line con 0 linevty 0 4 login linevty 5 15 login ! end

Configuration for Switch SW2

```
SW2#sh run
Building configuration...
Current configuration : 2082 bytes
!
version 12.2
no service pad
service timestamps debug datetimemsec
service timestamps log datetimemsec
no service password-encryption
!
hostname SW2
!
interface GigabitEthernet1/0/4
switchport trunk encapsulation dot1q
switchport mode trunk
!
!
interface GigabitEthernet1/0/5
switchport access vlan 10
switchport mode access
!
interface GigabitEthernet1/0/6
switchport access vlan 10
switchport mode access
!
I
interface GigabitEthernet1/0/8
switchport access vlan 10
!
Т
interface Vlan10
```

Capstone Project Report MINT-709 Failover solution for Elastix server over WAN

ip address 172.16.2.3 255.255.255.0 noip route-cache standby 10 ip 172.16.2.1 standby 10 timers 2 4 standby 10 priority 90 !

All the ports shown above belong to Vlan 10.

```
Configuration for NAT Router (Cisco 2600)
```

hostnameNAT_Router boot-start-marker boot-end-marker ! ! memory-sizeiomem 10 noaaa new-model ip subnet-zero ļ l l ipcef L L L L interface FastEthernet0/0 ip address 192.168.10.99 255.255.255.0 ipnat outside duplex auto speed auto ! interface Serial0/0 noip address shutdown ! interface FastEthernet0/1

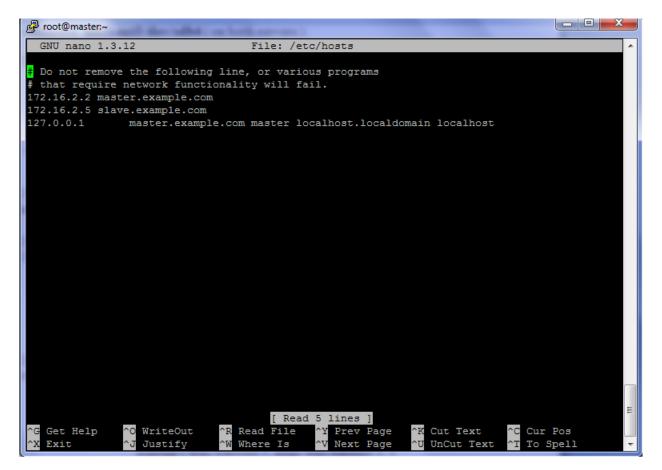
Capstone Project Report MINT-709 Failover solution for Elastix server over WAN

```
ip address 172.16.2.20 255.255.255.0
ipnat inside
duplex auto
speed auto
!
interface Serial0/1
noip address
shutdown
I
ipnat inside source list 1 interface FastEthernet0/0 overload
ip http server
ip classless
!
!
access-list 1 permit 172.16.2.0 0.0.0.255
!
l
I
voice-port 1/0/0
I
voice-port 1/0/1
l
ļ
I
!
line con 0
line aux 0
linevty 04
!
!
end
```

Note: Router R1, Router R4, NAT Router, Switch SW1 and Switch SW2 are at customer end.

In this test we need to make few changes during the installation of servers. We have to follow all the steps which are explained above except the following steps.

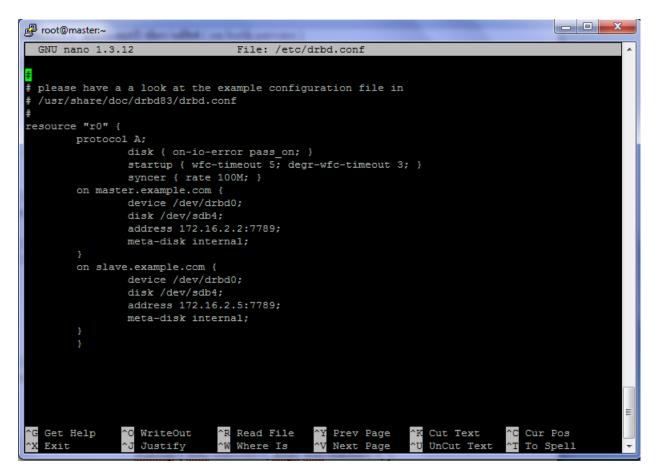
14. We have to edit /etc/hosts



Note: This should be same on both the servers.

Here we have defined that the server with ip address 172.16.2.2 has a hostname master.example.com and a server with ip address 172.16.2.5 has a hostname slave.example.com respectively and both are the nodes of the same cluster.

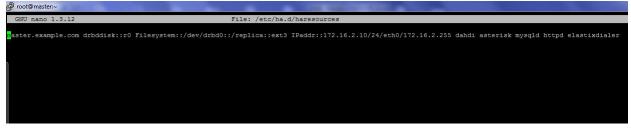
15. We need to change /etc/drbd.conf



Note: This should be same on both the servers.

Here we are defining the memory space which is to be synchronized between the servers with their host names. We have also defined the synchronization rate in this file, which in our case is 100 M.

27. We have to edit /etc/ha.d/haresources



Note: This should be same on both the servers.

In this file we have defined a floating ip address which is 172.16.2.10 and all the services which we want to run on both the servers.

28. In this step we will create a file that we want to have for high availability clustering, which has the authentication key defined in it.

🖉 root@master:~ GNU nano 1.3.12 File: /etc/ha.d/authkeys uth 1 shal elastix Get Help ^0 WriteOut ^R Read File ^Y Prev Page `K Cut Text Cur Pos Exit Justifv Where Is Next Page UnCut Text To Spell

Command: nano /etc/ha.d/authkeys

Split-Brain

If DRBD detects that both nodes are (or were at some point, while disconnected) in the primary role, it immediately tears down the replication connection. This situation is called Split-Brain. Split-Brain issue can be resolved manually as well as by running an automatic script. In our case we are going to do it manually. Following is the procedure for resolving an issue of split-brain manually.

If we get the following output, after executing drbdadm role r0 or if a server was taken out of service and then set it back on-line:

cs: StandAlonest:Secondary/Unknown ds:UpToDate/Unknown orcs: StandAlonest:Primary/Unknown ds:UpToDate/Unknown

To resolve this issue we have to execute following:

On out dated secondary server

- 1. drbdadm secondary r0
- 2. drbdadm disconnect all
- 3. drbdadm -- --discard-my-data connect r0

On Primary server

1. drbdadm connect r0

Now our servers are synchronized and are working as master and slave.

We have two ip phones, one is connected at the location where master server is located, with the name of TEST1@2001 with an ip addresss 172.16.2.30 and the other phone is connected on the other end with the name of TEST2@2002 with an ip address of 172.16.2.29.

Now it's time to test the behavior of the servers.

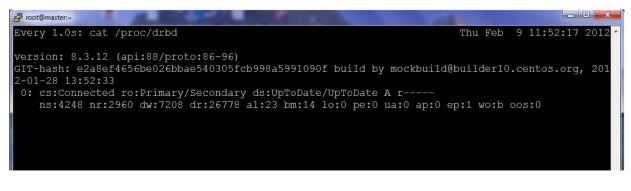
Final results

A. When both the servers are in the original state, which means heart beat is running between them and servers are fully synchronized with each other.

We can check the role and the synchronization status of the servers by executing

watch -n 1 cat /proc/drbd

Output on master server:-



In above screen shot we can see that both the servers are up to date and are acting normally.

Now we will make a call from TEST1@2001 to Test2@2002.

Output on server with ip address 172.16.2.10, which is our virtual ip.

Asterisk 1.8.7.0, Copyright (C) 1999 - 2011 Digium, Inc. and others. Created by Mark Spencer <markster@digium.com> Asterisk comes with ABSOLUTELY NO WARRANTY; type 'core show warranty' for details. This is free software, with components licensed under the GNU General Public License version 2 and other licenses; you are welcome to redistribute it under certain conditions. Type 'core show license' for details. _____ == Parsing '/etc/asterisk/asterisk.conf': == Found == Parsing '/etc/asterisk/extconfig.conf': == Found Connected to Asterisk 1.8.7.0 currently running on master (pid = 3982) Verbosity is at least 21 master*CLI> [2]+ Stopped asterisk -rvvvvvvvvvvvvvvvvvvvvv [root@master ~]# clear Asterisk 1.8.7.0, Copyright (C) 1999 - 2011 Digium, Inc. and others. Created by Mark Spencer <markster@digium.com> Asterisk comes with ABSOLUTELY NO WARRANTY; type 'core show warranty' for details. This is free software, with components licensed under the GNU General Public License version 2 and other licenses; you are welcome to redistribute it under certain conditions. Type 'core show license' for details.

== Parsing '/etc/asterisk/asterisk.conf': == Found

== Parsing '/etc/asterisk/extconfig.conf': == Found

Connected to Asterisk 1.8.7.0 currently running on master (pid = 3982) Verbosity is at least 21

== Using SIP RTP TOS bits 184

== Using SIP RTP CoS mark 5

-- Executing [2002@from-internal:1] Macro("SIP/2001-0000000e", "exten-

vm,2002,2002") in new stack

-- Executing [s@macro-exten-vm:1] Macro("SIP/2001-0000000e", "user-callerid,") in new stack

-- Executing [s@macro-user-callerid:1] Set("SIP/2001-0000000e", "AMPUSER=2001") in new stack

-- Executing [s@macro-user-callerid:2] Gotolf("SIP/2001-0000000e", "0?report") in new stack

-- Executing [s@macro-user-callerid:3] ExecIf("SIP/2001-0000000e",

"1?Set(REALCALLERIDNUM=2001)") in new stack

-- Executing [s@macro-user-callerid:4] Set("SIP/2001-0000000e", "AMPUSER=2001") in new stack

-- Executing [s@macro-user-callerid:5] Set("SIP/2001-0000000e",

"AMPUSERCIDNAME=Test1") in new stack

-- Executing [s@macro-user-callerid:6] Gotolf("SIP/2001-0000000e", "0?report") in new stack

-- Executing [s@macro-user-callerid:7] Set("SIP/2001-0000000e",

"AMPUSERCID=2001") in new stack

-- Executing [s@macro-user-callerid:8] Set("SIP/2001-0000000e",

"CALLERID(all)="Test1" <2001>") in new stack

-- Executing [s@macro-user-callerid:9] ExecIf("SIP/2001-0000000e",

"0?Set(CHANNEL(language)=)") in new stack

-- Executing [s@macro-user-callerid:10] Gotolf("SIP/2001-0000000e", "0?continue") in new stack

-- Executing [s@macro-user-callerid:11] Set("SIP/2001-0000000e", "__TTL=64") in new stack

-- Executing [s@macro-user-callerid:12] Gotolf("SIP/2001-0000000e", "1?continue") in new stack

-- Goto (macro-user-callerid,s,19)

-- Executing [s@macro-user-callerid:19] Set("SIP/2001-0000000e",

"CALLERID(number)=2001") in new stack

-- Executing [s@macro-user-callerid:20] Set("SIP/2001-0000000e",

"CALLERID(name)=Test1") in new stack

-- Executing [s@macro-user-callerid:21] NoOp("SIP/2001-0000000e", "Using CallerID "Test1" <2001>") in new stack -- Executing [s@macro-exten-vm:2] Set("SIP/2001-0000000e",

"RingGroupMethod=none") in new stack

-- Executing [s@macro-exten-vm:3] Set("SIP/2001-0000000e", "VMBOX=2002") in new stack

-- Executing [s@macro-exten-vm:4] Set("SIP/2001-0000000e", "__EXTTOCALL=2002") in new stack

-- Executing [s@macro-exten-vm:5] Set("SIP/2001-0000000e", "CFUEXT=") in new stack

-- Executing [s@macro-exten-vm:6] Set("SIP/2001-0000000e", "CFBEXT=") in new stack

-- Executing [s@macro-exten-vm:7] Set("SIP/2001-0000000e", "RT=15") in new stack -- Executing [s@macro-exten-vm:8] Macro("SIP/2001-0000000e", "record-

enable,2002,IN") in new stack

-- Executing [s@macro-record-enable:1] Gotolf("SIP/2001-0000000e", "1?check") in new stack

-- Goto (macro-record-enable,s,4)

-- Executing [s@macro-record-enable:4] ExecIf("SIP/2001-0000000e",

"0?MacroExit()") in new stack

-- Executing [s@macro-record-enable:5] Gotolf("SIP/2001-0000000e", "0?Group:OUT") in new stack

-- Goto (macro-record-enable,s,15)

-- Executing [s@macro-record-enable:15] Gotolf("SIP/2001-0000000e", "1?IN") in new stack

-- Goto (macro-record-enable,s,20)

-- Executing [s@macro-record-enable:20] ExecIf("SIP/2001-0000000e",

"1?MacroExit()") in new stack

-- Executing [s@macro-exten-vm:9] Macro("SIP/2001-0000000e", "dial-

one,15,tr,2002") in new stack

-- Executing [s@macro-dial-one:1] Set("SIP/2001-0000000e", "DEXTEN=2002") in new stack

-- Executing [s@macro-dial-one:2] Set("SIP/2001-0000000e", "DIALSTATUS_CW=") in new stack

-- Executing [s@macro-dial-one:3] GosubIf("SIP/2001-0000000e", "0?screen,1") in new stack

-- Executing [s@macro-dial-one:4] GosubIf("SIP/2001-0000000e", "0?cf,1") in new stack

-- Executing [s@macro-dial-one:5] Gotolf("SIP/2001-0000000e", "1?skip1") in new stack

-- Goto (macro-dial-one,s,8)

-- Executing [s@macro-dial-one:8] GotoIf("SIP/2001-0000000e", "0?nodial") in new stack

-- Executing [s@macro-dial-one:9] Gotolf("SIP/2001-0000000e", "0?continue") in new stack

-- Executing [s@macro-dial-one:10] Set("SIP/2001-0000000e", "EXTHASCW=") in new stack

-- Executing [s@macro-dial-one:11] GotoIf("SIP/2001-0000000e",

"1?next1:cwinusebusy") in new stack

-- Goto (macro-dial-one,s,12)

-- Executing [s@macro-dial-one:12] GotoIf("SIP/2001-0000000e", "0?docfu:skip3") in new stack

-- Goto (macro-dial-one,s,16)

-- Executing [s@macro-dial-one:16] Gotolf("SIP/2001-0000000e", "1?next2:continue") in new stack

-- Goto (macro-dial-one,s,17)

-- Executing [s@macro-dial-one:17] Gotolf("SIP/2001-0000000e", "1?continue") in new stack

-- Goto (macro-dial-one,s,25)

-- Executing [s@macro-dial-one:25] Gotolf("SIP/2001-0000000e", "0?nodial") in new stack

-- Executing [s@macro-dial-one:26] Gosublf("SIP/2001-0000000e",

"1?dstring,1:dlocal,1") in new stack

-- Executing [dstring@macro-dial-one:1] Set("SIP/2001-0000000e", "DSTRING=") in new stack

-- Executing [dstring@macro-dial-one:2] Set("SIP/2001-0000000e", "DEVICES=2002") in new stack

-- Executing [dstring@macro-dial-one:3] ExecIf("SIP/2001-0000000e", "0?Return()") in new stack

-- Executing [dstring@macro-dial-one:4] Execlf("SIP/2001-0000000e",

"0?Set(DEVICES=002)") in new stack

-- Executing [dstring@macro-dial-one:5] Set("SIP/2001-0000000e", "LOOPCNT=1") in new stack

-- Executing [dstring@macro-dial-one:6] Set("SIP/2001-0000000e", "ITER=1") in new stack

-- Executing [dstring@macro-dial-one:7] Set("SIP/2001-0000000e",

"THISDIAL=SIP/2002") in new stack

-- Executing [dstring@macro-dial-one:8] Gosublf("SIP/2001-0000000e",

"1?zap2dahdi,1") in new stack

-- Executing [zap2dahdi@macro-dial-one:1] ExecIf("SIP/2001-0000000e", "0?Return()") in new stack

-- Executing [zap2dahdi@macro-dial-one:2] Set("SIP/2001-0000000e", "NEWDIAL=") in new stack

-- Executing [zap2dahdi@macro-dial-one:3] Set("SIP/2001-0000000e",

"LOOPCNT2=1") in new stack

-- Executing [zap2dahdi@macro-dial-one:4] Set("SIP/2001-0000000e", "ITER2=1") in new stack

-- Executing [zap2dahdi@macro-dial-one:5] Set("SIP/2001-0000000e", "THISPART2=SIP/2002") in new stack

-- Executing [zap2dahdi@macro-dial-one:6] ExecIf("SIP/2001-0000000e", "0?Set(THISPART2=DAHDI/2002)") in new stack

-- Executing [zap2dahdi@macro-dial-one:7] Set("SIP/2001-0000000e", "NEWDIAL=SIP/2002&") in new stack

-- Executing [zap2dahdi@macro-dial-one:8] Set("SIP/2001-0000000e", "ITER2=2") in new stack

-- Executing [zap2dahdi@macro-dial-one:9] GotoIf("SIP/2001-0000000e", "0?begin2") in new stack

-- Executing [zap2dahdi@macro-dial-one:10] Set("SIP/2001-0000000e", "THISDIAL=SIP/2002") in new stack

-- Executing [zap2dahdi@macro-dial-one:11] Return("SIP/2001-0000000e", "") in new stack

-- Executing [dstring@macro-dial-one:9] Set("SIP/2001-0000000e",

"DSTRING=SIP/2002&") in new stack

-- Executing [dstring@macro-dial-one:10] Set("SIP/2001-0000000e", "ITER=2") in new stack

-- Executing [dstring@macro-dial-one:11] Gotolf("SIP/2001-0000000e", "0?begin") in new stack

-- Executing [dstring@macro-dial-one:12] Set("SIP/2001-0000000e", "DSTRING=SIP/2002") in new stack

-- Executing [dstring@macro-dial-one:13] Return("SIP/2001-0000000e", "") in new

stack

-- Executing [s@macro-dial-one:27] Gotolf("SIP/2001-0000000e", "0?nodial") in new stack

-- Executing [s@macro-dial-one:28] Gotolf("SIP/2001-0000000e", "1?skiptrace") in new stack

-- Goto (macro-dial-one,s,30)

-- Executing [s@macro-dial-one:30] Set("SIP/2001-0000000e", "D_OPTIONS=tr") in new stack

-- Executing [s@macro-dial-one:31] ExecIf("SIP/2001-0000000e",

"0?SIPAddHeader(Alert-Info:)") in new stack

-- Executing [s@macro-dial-one:32] ExecIf("SIP/2001-0000000e", "0?SIPAddHeader()") in new stack

-- Executing [s@macro-dial-one:33] ExecIf("SIP/2001-0000000e",

"0?Set(CHANNEL(musicclass)=)") in new stack

-- Executing [s@macro-dial-one:34] GosubIf("SIP/2001-0000000e", "0?qwait,1") in new stack

-- Executing [s@macro-dial-one:35] Set("SIP/2001-0000000e", "__CWIGNORE=") in new stack

-- Executing [s@macro-dial-one:36] Set("SIP/2001-0000000e", "__KEEPCID=TRUE") in new stack

-- Executing [s@macro-dial-one:37] Dial("SIP/2001-0000000e", "SIP/2002,15,tr") in new stack

== Using SIP RTP TOS bits 184

== Using SIP RTP CoS mark 5

-- Called SIP/2002

-- SIP/2002-0000000f is ringing

-- SIP/2002-0000000f is making progress passing it to SIP/2001-0000000e

-- SIP/2002-0000000f is ringing

-- SIP/2002-0000000f is making progress passing it to SIP/2001-0000000e

-- SIP/2002-0000000f answered SIP/2001-0000000e

-- Remote UNIX connection

-- Remote UNIX connection disconnected

-- Executing [h@macro-dial-one:1] Macro("SIP/2001-0000000e", "hangupcall,") in new stack

-- Executing [s@macro-hangupcall:1] Gotolf("SIP/2001-0000000e",

"1?endmixmoncheck") in new stack

-- Goto (macro-hangupcall,s,9)

-- Executing [s@macro-hangupcall:9] NoOp("SIP/2001-0000000e", "End of MIXMON check") in new stack

-- Executing [s@macro-hangupcall:10] Gotolf("SIP/2001-0000000e",

"1?nomeetmemon") in new stack

-- Goto (macro-hangupcall,s,15)

-- Executing [s@macro-hangupcall:15] NoOp("SIP/2001-0000000e",

"MEETME_RECORDINGFILE=") in new stack

-- Executing [s@macro-hangupcall:16] Gotolf("SIP/2001-0000000e", "1?noautomon") in new stack

-- Goto (macro-hangupcall,s,18)

-- Executing [s@macro-hangupcall:18] NoOp("SIP/2001-0000000e",

"TOUCH_MONITOR_OUTPUT=") in new stack

-- Executing [s@macro-hangupcall:19] Gotolf("SIP/2001-0000000e", "1?noautomon2") in new stack

-- Goto (macro-hangupcall,s,25)

-- Executing [s@macro-hangupcall:25] NoOp("SIP/2001-0000000e",

"MONITOR_FILENAME=") in new stack

-- Executing [s@macro-hangupcall:26] GotoIf("SIP/2001-0000000e", "1?skiprg") in new stack

-- Goto (macro-hangupcall,s,29)

-- Executing [s@macro-hangupcall:29] Gotolf("SIP/2001-0000000e", "1?skipblkvm") in new stack

-- Goto (macro-hangupcall,s,32)

-- Executing [s@macro-hangupcall:32] Gotolf("SIP/2001-0000000e", "1?theend") in new stack

-- Goto (macro-hangupcall,s,34)

-- Executing [s@macro-hangupcall:34] Hangup("SIP/2001-0000000e", "") in new stack == Spawn extension (macro-hangupcall, s, 34) exited non-zero on 'SIP/2001-0000000e' in macro 'hangupcall'

== Spawn extension (macro-dial-one, h, 1) exited non-zero on 'SIP/2001-0000000e'

== Spawn extension (macro-dial-one, s, 37) exited non-zero on 'SIP/2001-0000000e' in macro 'dial-one'

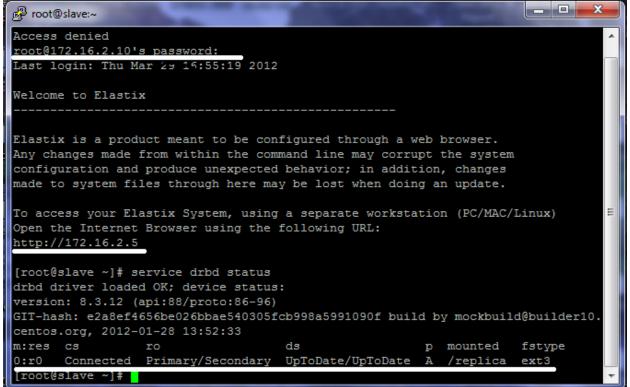
== Spawn extension (macro-exten-vm, s, 9) exited non-zero on 'SIP/2001-0000000e' in macro 'exten-vm'

== Spawn extension (from-internal, 2002, 1) exited non-zero on 'SIP/2001-0000000e'

The highlighted lines show that the call is in progress.

B. Now it's time to check what will happen when heartbeat stops on master server. Execute this on master server: Service heartbeat stop

Output on the server with ip address 172.16.2.10 (Floating ip address) :-



In the above screen shot we are logged in to the server through our floating ip address which is 172.16.2.10. As the heartbeat service is stopped manually between the two servers, drbd status changes. Our secondary server automatically changes to primary server. Now our server with ip address 172.16.2.5 is acting like a primary server and the server with the ip address 172.16.2.2 is a secondary server. All the phones are registered with the floating ip address.

We will again make a call from TEST1@2001 to Test2@2002.

Output:

[root@slave ~]# asterisk -rvvvvvvvvvvvvvvvvvvvvv

Asterisk 1.8.7.0, Copyright (C) 1999 - 2011 Digium, Inc. and others.

Created by Mark Spencer <markster@digium.com>

Asterisk comes with ABSOLUTELY NO WARRANTY; type 'core show warranty' for detail s.

This is free software, with components licensed under the GNU General Public License version 2 and other licenses; you are welcome to redistribute it under certain conditions. Type 'core show license' for details.

== Parsing '/etc/asterisk/asterisk.conf': == Found

== Parsing '/etc/asterisk/extconfig.conf': == Found

Connected to Asterisk 1.8.7.0 currently running on slave (pid = 4021) Verbosity is at least 17

-- Executing [h@macro-dial-one:1] Macro("SIP/2001-00000000", "hangupcall,") in new stack

-- Executing [s@macro-hangupcall:1] Gotolf("SIP/2001-00000000",

"1?endmixmoncheck") in new stack

-- Goto (macro-hangupcall,s,9)

-- Executing [s@macro-hangupcall:9] NoOp("SIP/2001-00000000", "End of MIXMON check") in new stack

-- Executing [s@macro-hangupcall:10] Gotolf("SIP/2001-00000000",

"1?nomeetmemon") in new stack

-- Goto (macro-hangupcall,s,15)

-- Executing [s@macro-hangupcall:15] NoOp("SIP/2001-00000000",

"MEETME_RECORDINGFILE=") in new stack

-- Executing [s@macro-hangupcall:16] Gotolf("SIP/2001-00000000", "1?noautomon") in new stack

-- Goto (macro-hangupcall,s,18)

-- Executing [s@macro-hangupcall:18] NoOp("SIP/2001-00000000",

"TOUCH_MONITOR_OUTPUT=") in new stack

-- Executing [s@macro-hangupcall:19] Gotolf("SIP/2001-00000000", "1?noautomon2") in new stack

-- Goto (macro-hangupcall,s,25)

-- Executing [s@macro-hangupcall:25] NoOp("SIP/2001-00000000", "MONITOR FILENAME=") in new stack

-- Executing [s@macro-hangupcall:26] Gotolf("SIP/2001-00000000", "1?skiprg") in new stack

-- Goto (macro-hangupcall,s,29)

-- Executing [s@macro-hangupcall:29] Gotolf("SIP/2001-00000000", "1?skipblkvm") in new stack

-- Goto (macro-hangupcall,s,32)

-- Executing [s@macro-hangupcall:32] Gotolf("SIP/2001-00000000", "1?theend") in new stack

-- Goto (macro-hangupcall,s,34)

-- Executing [s@macro-hangupcall:34] Hangup("SIP/2001-00000000", "") in new stack

== Spawn extension (macro-hangupcall, s, 34) exited non-zero on 'SIP/2001-00000000' in macro 'hangupcall'

== Spawn extension (macro-dial-one, h, 1) exited non-zero on 'SIP/2001-00000000'

== Spawn extension (macro-dial-one, s, 37) exited non-zero on 'SIP/2001-00000000' in macro 'dial-one'

== Spawn extension (macro-exten-vm, s, 9) exited non-zero on 'SIP/2001-00000000' in macro 'exten-vm'

== Spawn extension (from-internal, 2002, 1) exited non-zero on 'SIP/2001-00000000' == Using SIP RTP TOS bits 184

== Using SIP RTP CoS mark 5

-- Executing [2002@from-internal:1] Macro("SIP/2001-00000002", "exten-

vm,2002,2002") in new stack

-- Executing [s@macro-exten-vm:1] Macro("SIP/2001-00000002", "user-callerid,") in new stack

-- Executing [s@macro-user-callerid:1] Set("SIP/2001-00000002", "AMPUSER=2001") in new stack

-- Executing [s@macro-user-callerid:2] Gotolf("SIP/2001-00000002", "0?report") in new stack

-- Executing [s@macro-user-callerid:3] ExecIf("SIP/2001-0000002",

"1?Set(REALCALLERIDNUM=2001)") in new stack

-- Executing [s@macro-user-callerid:4] Set("SIP/2001-00000002", "AMPUSER=2001") in new stack

-- Executing [s@macro-user-callerid:5] Set("SIP/2001-00000002",

"AMPUSERCIDNAME=Test1") in new stack

-- Executing [s@macro-user-callerid:6] Gotolf("SIP/2001-00000002", "0?report") in new stack

-- Executing [s@macro-user-callerid:7] Set("SIP/2001-00000002", "AMPUSERCID=2001") in new stack

-- Executing [s@macro-user-callerid:8] Set("SIP/2001-00000002",

"CALLERID(all)="Test1" <2001>") in new stack

-- Executing [s@macro-user-callerid:9] Execlf("SIP/2001-00000002", "0?Set(CHANNEL(language)=)") in new stack

-- Executing [s@macro-user-callerid:10] GotoIf("SIP/2001-00000002", "0?continue") in new stack

-- Executing [s@macro-user-callerid:11] Set("SIP/2001-00000002", "__TTL=64") in new stack

-- Executing [s@macro-user-callerid:12] Gotolf("SIP/2001-00000002", "1?continue") in new stack

-- Goto (macro-user-callerid,s,19)

-- Executing [s@macro-user-callerid:19] Set("SIP/2001-00000002", "CALLERID(number)=2001") in new stack

-- Executing [s@macro-user-callerid:20] Set("SIP/2001-0000002",

"CALLERID(name)=Test1") in new stack

-- Executing [s@macro-user-callerid:21] NoOp("SIP/2001-00000002", "Using CallerID "Test1" <2001>") in new stack

-- Executing [s@macro-exten-vm:2] Set("SIP/2001-00000002",

"RingGroupMethod=none") in new stack

-- Executing [s@macro-exten-vm:3] Set("SIP/2001-00000002", "VMBOX=2002") in new stack

-- Executing [s@macro-exten-vm:4] Set("SIP/2001-00000002", "__EXTTOCALL=2002") in new stack

-- Executing [s@macro-exten-vm:5] Set("SIP/2001-00000002", "CFUEXT=") in new stack

-- Executing [s@macro-exten-vm:6] Set("SIP/2001-00000002", "CFBEXT=") in new stack

-- Executing [s@macro-exten-vm:7] Set("SIP/2001-00000002", "RT=15") in new stack

-- Executing [s@macro-exten-vm:8] Macro("SIP/2001-00000002", "recordenable,2002,IN") in new stack

-- Executing [s@macro-record-enable:1] Gotolf("SIP/2001-00000002", "1?check") in new stack

-- Goto (macro-record-enable,s,4)

-- Executing [s@macro-record-enable:4] ExecIf("SIP/2001-00000002",

"0?MacroExit()") in new stack

-- Executing [s@macro-record-enable:5] Gotolf("SIP/2001-00000002",

"0?Group:OUT") in new stack

-- Goto (macro-record-enable,s,15)

-- Executing [s@macro-record-enable:15] Gotolf("SIP/2001-00000002", "1?IN") in new stack

-- Goto (macro-record-enable,s,20)

-- Executing [s@macro-record-enable:20] ExecIf("SIP/2001-00000002",

"1?MacroExit()") in new stack

-- Executing [s@macro-exten-vm:9] Macro("SIP/2001-00000002", "dialone,15,tr,2002") in new stack

-- Executing [s@macro-dial-one:1] Set("SIP/2001-00000002", "DEXTEN=2002") in new stack

-- Executing [s@macro-dial-one:2] Set("SIP/2001-00000002", "DIALSTATUS_CW=") in new stack

-- Executing [s@macro-dial-one:3] GosubIf("SIP/2001-00000002", "0?screen,1") in new stack

-- Executing [s@macro-dial-one:4] Gosublf("SIP/2001-00000002", "0?cf,1") in new stack

-- Executing [s@macro-dial-one:5] Gotolf("SIP/2001-00000002", "1?skip1") in new stack

-- Goto (macro-dial-one,s,8)

-- Executing [s@macro-dial-one:8] Gotolf("SIP/2001-00000002", "0?nodial") in new stack

-- Executing [s@macro-dial-one:9] Gotolf("SIP/2001-00000002", "0?continue") in new stack

-- Executing [s@macro-dial-one:10] Set("SIP/2001-00000002", "EXTHASCW=") in new stack

-- Executing [s@macro-dial-one:11] GotoIf("SIP/2001-0000002",

"1?next1:cwinusebusy") in new stack

-- Goto (macro-dial-one,s,12)

-- Executing [s@macro-dial-one:12] GotoIf("SIP/2001-00000002", "0?docfu:skip3") in new stack

-- Goto (macro-dial-one,s,16)

-- Executing [s@macro-dial-one:16] Gotolf("SIP/2001-00000002", "1?next2:continue") in new stack

-- Goto (macro-dial-one,s,17)

-- Executing [s@macro-dial-one:17] Gotolf("SIP/2001-00000002", "1?continue") in new stack

-- Goto (macro-dial-one,s,25)

-- Executing [s@macro-dial-one:25] GotoIf("SIP/2001-00000002", "0?nodial") in new stack

-- Executing [s@macro-dial-one:26] Gosublf("SIP/2001-00000002",

"1?dstring,1:dlocal,1") in new stack

-- Executing [dstring@macro-dial-one:1] Set("SIP/2001-00000002", "DSTRING=") in new stack

-- Executing [dstring@macro-dial-one:2] Set("SIP/2001-00000002", "DEVICES=2002") in new stack

-- Executing [dstring@macro-dial-one:3] ExecIf("SIP/2001-00000002", "0?Return()") in new stack

-- Executing [dstring@macro-dial-one:4] Execlf("SIP/2001-00000002", "0?Set(DEVICES=002)") in new stack

-- Executing [dstring@macro-dial-one:5] Set("SIP/2001-00000002", "LOOPCNT=1") in new stack

-- Executing [dstring@macro-dial-one:6] Set("SIP/2001-00000002", "ITER=1") in new stack

-- Executing [dstring@macro-dial-one:7] Set("SIP/2001-00000002",

"THISDIAL=SIP/2002") in new stack

-- Executing [dstring@macro-dial-one:8] GosubIf("SIP/2001-0000002",

"1?zap2dahdi,1") in new stack

-- Executing [zap2dahdi@macro-dial-one:1] ExecIf("SIP/2001-00000002", "0?Return()") in new stack

-- Executing [zap2dahdi@macro-dial-one:2] Set("SIP/2001-00000002", "NEWDIAL=") in new stack

-- Executing [zap2dahdi@macro-dial-one:3] Set("SIP/2001-00000002",

"LOOPCNT2=1") in new stack

-- Executing [zap2dahdi@macro-dial-one:4] Set("SIP/2001-00000002", "ITER2=1") in new stack

-- Executing [zap2dahdi@macro-dial-one:5] Set("SIP/2001-0000002", "THISPART2=SIP/2002") in new stack

-- Executing [zap2dahdi@macro-dial-one:6] ExecIf("SIP/2001-00000002", "0?Set(THISPART2=DAHDI/2002)") in new stack

-- Executing [zap2dahdi@macro-dial-one:7] Set("SIP/2001-00000002", "NEWDIAL=SIP/2002&") in new stack

-- Executing [zap2dahdi@macro-dial-one:8] Set("SIP/2001-00000002", "ITER2=2") in new stack

-- Executing [zap2dahdi@macro-dial-one:9] GotoIf("SIP/2001-00000002", "0?begin2") in new stack

-- Executing [zap2dahdi@macro-dial-one:10] Set("SIP/2001-00000002", "THISDIAL=SIP/2002") in new stack

-- Executing [zap2dahdi@macro-dial-one:11] Return("SIP/2001-00000002", "") in new stack

-- Executing [dstring@macro-dial-one:9] Set("SIP/2001-00000002",

"DSTRING=SIP/2002&") in new stack

-- Executing [dstring@macro-dial-one:10] Set("SIP/2001-00000002", "ITER=2") in new stack

-- Executing [dstring@macro-dial-one:11] Gotolf("SIP/2001-00000002", "0?begin") in new stack

-- Executing [dstring@macro-dial-one:12] Set("SIP/2001-00000002", "DSTRING=SIP/2002") in new stack -- Executing [dstring@macro-dial-one:13] Return("SIP/2001-00000002", "") in new stack

-- Executing [s@macro-dial-one:27] Gotolf("SIP/2001-00000002", "0?nodial") in new stack

-- Executing [s@macro-dial-one:28] Gotolf("SIP/2001-00000002", "1?skiptrace") in new stack

-- Goto (macro-dial-one,s,30)

-- Executing [s@macro-dial-one:30] Set("SIP/2001-00000002", "D_OPTIONS=tr") in new stack

-- Executing [s@macro-dial-one:31] Execlf("SIP/2001-00000002", "0?SIPAddHeader(Alert-Info:)") in new stack

-- Executing [s@macro-dial-one:32] ExecIf("SIP/2001-00000002", "0?SIPAddHeader()") in new stack

-- Executing [s@macro-dial-one:33] ExecIf("SIP/2001-0000002",

"0?Set(CHANNEL(musicclass)=)") in new stack

-- Executing [s@macro-dial-one:34] GosubIf("SIP/2001-00000002", "0?qwait,1") in new stack

-- Executing [s@macro-dial-one:35] Set("SIP/2001-00000002", "__CWIGNORE=") in new stack

-- Executing [s@macro-dial-one:36] Set("SIP/2001-00000002", "__KEEPCID=TRUE") in new stack

-- Executing [s@macro-dial-one:37] Dial("SIP/2001-00000002", "SIP/2002,15,tr") in new stack

== Using SIP RTP TOS bits 184

== Using SIP RTP CoS mark 5

-- Called SIP/2002

-- SIP/2002-0000003 is ringing

-- SIP/2002-00000003 is making progress passing it to SIP/2001-00000002

-- SIP/2002-0000003 is ringing

-- SIP/2002-00000003 is making progress passing it to SIP/2001-00000002

-- SIP/2002-00000003 answered SIP/2001-00000002

-- Executing [h@macro-dial-one:1] Macro("SIP/2001-00000002", "hangupcall,") in new stack

-- Executing [s@macro-hangupcall:1] Gotolf("SIP/2001-0000002",

"1?endmixmoncheck") in new stack

-- Goto (macro-hangupcall,s,9)

-- Executing [s@macro-hangupcall:9] NoOp("SIP/2001-00000002", "End of MIXMON check") in new stack

-- Executing [s@macro-hangupcall:10] GotoIf("SIP/2001-0000002",

"1?nomeetmemon") in new stack

-- Goto (macro-hangupcall,s,15)

-- Executing [s@macro-hangupcall:15] NoOp("SIP/2001-00000002", "MEETME RECORDINGFILE=") in new stack

-- Executing [s@macro-hangupcall:16] Gotolf("SIP/2001-00000002", "1?noautomon") in new stack

-- Goto (macro-hangupcall,s,18)

-- Executing [s@macro-hangupcall:18] NoOp("SIP/2001-0000002",

"TOUCH_MONITOR_OUTPUT=") in new stack

-- Executing [s@macro-hangupcall:19] GotoIf("SIP/2001-00000002", "1?noautomon2") in new stack

-- Goto (macro-hangupcall,s,25)

-- Executing [s@macro-hangupcall:25] NoOp("SIP/2001-00000002",

"MONITOR_FILENAME=") in new stack

-- Executing [s@macro-hangupcall:26] GotoIf("SIP/2001-00000002", "1?skiprg") in new stack

-- Goto (macro-hangupcall,s,29)

-- Executing [s@macro-hangupcall:29] Gotolf("SIP/2001-00000002", "1?skipblkvm") in new stack

-- Goto (macro-hangupcall,s,32)

-- Executing [s@macro-hangupcall:32] GotoIf("SIP/2001-00000002", "1?theend") in new stack

-- Goto (macro-hangupcall,s,34)

-- Executing [s@macro-hangupcall:34] Hangup("SIP/2001-00000002", "") in new stack == Spawn extension (macro-hangupcall, s, 34) exited non-zero on 'SIP/2001-00000002' in macro 'hangupcall'

== Spawn extension (macro-dial-one, h, 1) exited non-zero on 'SIP/2001-0000002'

== Spawn extension (macro-dial-one, s, 37) exited non-zero on 'SIP/2001-00000002' in macro 'dial-one'

== Spawn extension (macro-exten-vm, s, 9) exited non-zero on 'SIP/2001-00000002' in macro 'exten-vm'

== Spawn extension (from-internal, 2002, 1) exited non-zero on 'SIP/2001-00000002'

The highlighted lines show that the call is in progress.

Conclusion

Both the tests show that redundancy for the Elastix servers is feasible over the wide area network, but the only challenge was to keep the same network on the both servers. No matter where they are located. We can use L2TP or VPLS for that, depending on our requirements.

Challenges and difficulties

- In this project the biggest challenge was to learn about Elastix server due to the limited resources in English language. As Elastix is a new open source communication server there is not enough study material over the internet so far.
- I was new to all the technologies that I have used in this project but with the help of online tutorials, videos and books I was able to understand and implement VPLS, L2TP, High availability clustering and Heartbeat.

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