CONTACT DETAILS

EMAIL :ashiqvoip@gmail.com

PHONE : +91 9137372342

ADDRESS : Perfect Dormitory
Perfect Dormitory, Saki Naka,
Mumbai, Maharashtra

COURSES COMPLETED

▶ **CISCO Certified Network Associate (Routing & Switching)**

▶ **CISCO Certified Network Associate (CCNA Voice)**

▶ **CISCO Certified Network Professional (CCNP Voice)**

EDUCATION

▶ **B.E in Information science &Engineering (2011 - 2015)**

▶ **Plus Two, Board of Higher Secondary Examination (2009 - 2011)**

▶ **SSLC from Central Board of Secondary Education (2008)**

PERSONAL OVERVIEW

FATHER'S NAME : ALI M.T
GENDER : MALE
DOB : 06/05/1990
MARITAL STATUS : MARRIED
LANGUAGES : ENGLISH
NATIONALITY : INDIAN

MOHAMED ASHIQ
VOICE ENGINEER
8 YEARS OF TOTAL EXPERIENCE &
5+ YEARS OF RELEVANT EXPERIENCE

CAREER OBJECTIVE

Aspire to acquire a strong job position as a VoIP Engineer, where I can skillfully utilize my past experiences in the new working environment, to achieve the company goals and objectives.

PROFESSIONAL EXPERIENCE

▶ **Current Employer** : NTT INDIA PRIVATE LIMITED
Designation : M S Engineer L2 (VOICE)
Duration : Dec 2021 to Till date.

▶ **Previous Employer** : ALPHASERVE TECHNOLOGIES
Designation : Voice Network Engineer
Duration : July 2018 to Dec 2021.

Previous Employer : Adora Info Solutions
Designation : Network Service Engineer
Duration : Sep 2015 to June 2018

JOB RESPONSIBILITIES

- Moving, Adding and Deleting Extensions,SoftIP,Agents.
- Troubleshooting and providing support of cisco IPT for clients.
- Configuration of user features.
- Troubleshooting of IP Phone registration issues.
- Configuring using Bulk Administration Tool (BAT).
- Configuration of soft key template and phone button template.
- Monitoring & Troubleshooting of Voice of UCS Sever, SIP Trunks and Back up.
- Troubleshooting of voice gateway protocol.
- Configuration of Extension mobility.
- Daily-wise health check-up (CUCM, Gateway).
- Configuration and Troubleshooting of Hunt.
- Configuration of SRST and call signaling protocols on Voice Gateways.
- Configuring SNR for requested users.
- Alert Monitoring and Troubleshooting through RTMT and PCA. Analysing
- number using DNA.
- Class of Service using Partitions and CSS.

⚙ TECHNICAL SKILLS

- Configuration of various features in Cisco Unified Communications Manager (CUCM)
- Configuration of CME (Call Manager Express) on Cisco routers.
- Bulk Administration (BAT) of IP Phones and IP Communicator.
- User Features such as Speed dial, Call forwarding, Call Park.
- Phone Button Template, Soft key Template.
- Configuration of Device Pool, Region, Location.
- Gained solid understanding and knowledge on various VOIP protocols such as SIP, MGCP, H.323, and SCCP.
- Configuration of Cisco IP Phones and third party SIP phones.
- Configuring and troubleshooting of PRI (E1 and T1), ISDN, T1/E1 CAS and FX trunks.
- Configuration of VoIP & POTS dial-peers.
- Configuration of Voice Gateways (SIP, MGCP, H.323).
- Managing dial plan using Route Pattern, Route List, Route Group, Local-Route Group, Translation Pattern, CMC, FAC and SLRG.
- Class of Service and Class of restriction.
- Integrating Unity Connection with CUCM (SCCP and SIP integration).
- Configuration of Media resources such as; MOH, Conference Bridge, MTP and Transcoder.
- Configuration of Extension Mobility.
- Various monitoring tools like RTMT, DNA and various router debug and show commands.
- Configuration of CAC and AAR.
- Configuring Voice Mail for LDAP users.
- Configuring and maintaining SNR issues.
- Configuration and management of SRST and MGCP fallback.
- Provisioning of partitions and Calling Search Spaces (CSS).

📜 DECLARATION

I hereby declare that the above stated information is true to the best of my knowledge and belief.

MOHAMED ASHIQ