

Avaya Aura[™] Session Manager Overview

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Chapter 1: Overview

What is new in this release

Avaya Aura[™] Session Manager 6.0 provides following set of functionality:

- Fully SIP redundant core provides active-active N+1 redundancy with up to six Session Manager instances in an enterprise
- Support for upto 250 SIP survivable branches
- Enhanced SIP feature set and connects to any IETF compliant SIP solution.
- Support for mixed H.323, DCP, and SIP endpoints at the redundant core and at the survivable branch.
- New scalability upto 100,000 users overall, 50,000 SIP stations, and 25,000 SIP entities.
- Expanded centralized management with Avaya Aura[™] System Manager 6.0 and can handle up to 25,000 locations in a full deployment with up to 750,000 BHCC of interlocation calling.

The left navigation pane for Avaya Aura[™] System Manager 6.0 web console has changed and Session Manager functionality now falls under Elements menu. Network Routing Policy menu has been renamed as Routing and SIP Domains as Domains.

What is Session Manager?

Companies typically have a diverse set of communications products within their corporate intranet that cannot communicate with each other. A standard signaling protocol is required to make these products work together. Avaya has adopted the Session Initiation Protocol (SIP) as the signaling protocol for communication.

Avaya Aura[™]Session Manager is a SIP routing and integration tool and the core component within the Avaya Aura[™] Enterprise Edition solution. It integrates all the SIP entities across the entire enterprise network within a company. Session Manager offers a new perspective on enterprise communication where individual locations are no longer managed as separate units

within the enterprise. Each location, branch, or application is part of the overall enterprise, managed as an enterprise, and seen as an enterprise. Session Manager offers:

- a simplified network-wide feature deployment
- centralized routing, SIP trunking, and user profiles
- cost-effective scalability (from small to very large deployments)
- high availability with geographic redundancy
- a secure environment that conforms to specific SIP standards and practices

What does Session Manager do?

Session Manager offers a core communication service that builds on existing equipment but adds a SIP-based architecture.

Session Manager connects Avaya Aura[™] Communication Manager as a feature (SIP) server and both Avaya enterprise PBX and small key PBX systems within branch offices, third-party PBXs, gateways, service providers, SIP-enabled adjuncts, and SIP and non-SIP telephones. It also integrates locations and applications. Specifically, Session Manager

- Normalizes disparate networks
- Routes SIP sessions across the network
- Provides the gateway for the enterprise for external SIP trunking
- Integrates with third-party equipment and endpoints
- Offers centralized management, including user profiles, through System Manager
- Supports SIP survivable branches
- Communicates with a Session Border Controller and provides protection at the edge of the enterprise network

Each Session Manager installation combines several or all of the following configurations:

- Centralized routing and dial plan management
- · Policy-based routing
 - Time of day routing
 - Alternate routing
 - Load balancing
 - Call admission control
- Tail end hop off (TEHO)
- Centralized SIP trunking

- Centralized applications
- Sequenced applications

It also handles all call re-direction, internal network call accounting feeds, toll by-pass, interoffice routing, and international least-cost routing.

Related topics:

Normalization of disparate networks on page 9

Centralized routing and dial plan on page 9

Tail end hop off on page 11

Centralized SIP trunking on page 11

Centralized applications on page 12

Sequenced applications on page 12

Normalization of disparate networks

Session Manager normalizes and adapts disparate SIP protocols to meet the strict SIP standards of the network. This allows third-party PBXs to work with each other and with Avaya equipment, allowing customers to realize true vendor interoperability.

An example is the ability to connect Cisco and Nortel PBXs with Session Manager so they operate with each other and with Avaya equipment. Session Manager converts the headers in the SIP messages that are used to display calling and called-party information between the format required by each switch in a call.

Centralized routing and dial plan

Session Manager provides the centralized, global dial plan for an enterprise and access to external SIP trunking. The dial plan is managed through a centralized management console and governs PBXs that connect using SIP to one or more Session Manager instances. For example, an enterprise might have two instances of Session Manager in two different data centers with two different SIP trunking service providers.

Dual dial plan

Within Session Manager, dial plans are administered without requiring the various PBXs to change their dial plans or digit manipulations. Session Manager also does not require that the PBXs send or accept numbers that are unique across the enterprise, sometimes called the enterprise canonical form. This saves customers from having to re-administer or re-engineer each PBX.

Mixed, diverse enterprise

Many enterprises consist of several smaller dial plan domains that are a result of multiple mergers or acquisitions. Session Manager supports these enterprises and moves the administration of these various dial plans to the Session Manager core so that one single point

of management is possible. This not only eliminates the need to administer and maintain these various dial plans in each PBX, but it allows the users in these locations to maintain the status quo and eliminates the need for users to change their dialing patterns.

Policy-based routing

Customers can define their call routing policy with Session Manager. These policies allow customers to control when calls are made, how the call load is balanced, and how calls are routed during network failures.

Least-cost routing, also called time-of-day routing, chooses the lowest cost route from a list of service providers on a time-of-day or time-of-week basis. This results in cost savings for the enterprise. The routing table shows an example of setting up time-of-day routing.

Route #	Route Priority	D	Start Time 1	Host 1		Start Time 6	Host 6
		Mon	0:00	Service Provider 1		20:00	Service Provider 2
	1		0 0 0				0 0
		Sun	0:00	Service Provider 2		20:00	Service Provider 1
		Mon	0:00	Service Provider 3		18:00	Service Provider 4
1	2		0 0 0		000		o o o
		Sun	0:00	Service Provider 4		18:00	Service Provider 3
			0 0 0				000
		Mon	0:00	Service Provider 5		14:00	Service Provider 6
	6		0				0 0
		Sun	0:00	Service Provider 6		14:00	Service Provider 5

Alternate routing routes calls around network failures on a global basis and uses global PSTN fallback when the internal network is unavailable.

Load balancing distributes calls to a SIP entity between multiple IP addresses. It is necessary for Avaya Voice Portal MPPs, Avaya Modular Messaging MASs, and SIP entities that are routed to more than one Session Manager to create redundancy. Load balancing uses priority and weights when selecting routes. Session Manager can be administered to choose between multiple IP addresses for a given entity and select these hosts based upon administered priorities and weights. If some entries have the same priority, then for each priority level, Session Manager picks a host according to the specified weights, with a lower weight receiving a higher priority.

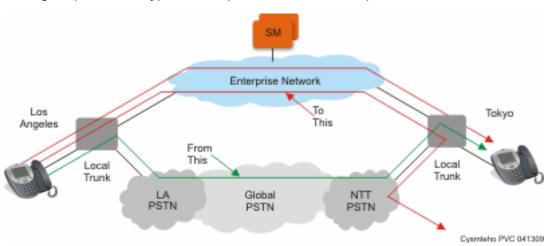
Call admission control allows calls to be rerouted when the WAN link to a branch fills up. Session Manager maintains the status of the link to each location, using administrator-set limits on each link.

Tail end hop off

Session Manager supports using the local trunks at each location to allow all users across the network enterprise to save toll charges for calls that go off the network. Outgoing calls are directed to local trunks on each location.

For example, calls from Tokyo to Los Angeles through a company's intranet are routed through the PSTN from the Los Angeles PBX, basically making it a *local* call from Los Angeles. And calls bound for Tokyo are routed through the Tokyo PBX. Local PBXs may not accept the full E.164 number, but Session Manager can change them to numbers that the PBXs can understand.

The figure provides a typical example of how tail end hop off works.



Centralized SIP trunking

Centralized SIP trunking allows SIP trunks to be shared within an enterprise. SIP trunks route calls from one PBX through a service provider over the network through a second service provider to another PBX. This requires each PBX to provide the SIP trunks. With Session Manager as the core, the individual PBXs share the SIP trunks. As a result, companies

- require fewer SIP trunks to provide the same level of service
- are able to buy SIP trunks in bulk
- eliminate local access costs, accessing the SIP trunking service provider directly.

Centralized applications

Session Manager provides connectivity for centralized Avaya applications such as Avaya Modular Messaging, Avaya Voice Portal, and Avaya Meeting Exchange. Each PBX, gateway, or location connects to Modular Messaging through the Session Manager core rather than individually. Session Manager also connects to SIP-enabled adjuncts, making the management and deployment of adjuncts infinitely simpler than the mesh-connect methods where each PBX connects to its own adjunct.

Sequenced applications

Sequenced applications are a series of applications that engage automatically. The sequence can be different or the same for each user. Each user is given a template of applications that is applied to every incoming, outgoing, or combined call for that user. Each application in a sequence sees all requests and can deny, modify, or forward initial SIP requests. Examples of sequenced Avaya applications are

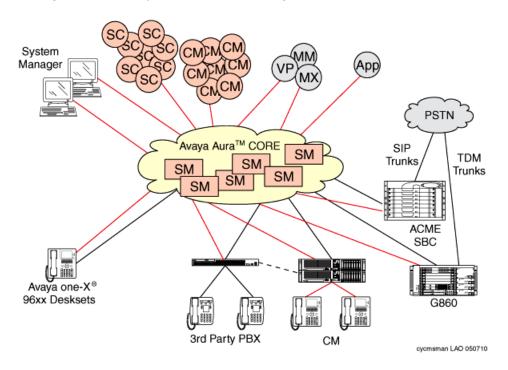
- Billing Service
- Voice Monitor
- Communication Manager Feature Server
- Call Blocker
- · Personal assistant
- Meeting Coordinator.

Session Manager also supports third-party PBX endpoint application sequencing. Because calls to and from users on non-Avaya PBXs are directed to the Session Manager core, applications are applied to calls to and from these endpoints. Session Manager creates a profile for third-party PBX users, and applies these applications to these users. Typical applications include blocking calls based on user preferences, directing calls to these users when they move across the Avaya Aura[™]-powered enterprise, and augmenting caller identification information for incoming and outgoing calls. This capability is available without upgrading or modifying code on existing third-party PBX equipment.

Session Manager architecture

Session Manager is the core of the network with SIP entities connecting to it. A grouping of a Session Manager with Avaya Aura[™] System Manager and SIP entities is called a Session Manager instance.

The figure shows a typical Session Manager instance with multiple SIP entities.



SM	Avaya Aura™ Session Manager
VP	Avaya Voice Portal 5.1 and 6.0
MM	Avaya Modular Messaging Release 6.0, 5.2 and 5.1.
MX	Avaya Meeting Exchange 6.0 and 5.1
G860	Avaya G860 High Density Trunk Gateway R2
SC	Survivable Core
Арр	other SIP applications
СМ	Communication Manager
ACME SBC	Acme Packets Session Border Controller 2900/3800/4500

A Session Manager instance consists of one server supporting up to 50,000 SIP entities. An enterprise network can support up to six Session Manager instances. These Session Manager

instances can be installed in the same data center or in multiple data centers and in geographically redundant locations with virtually unlimited distance restrictions. All of the instances support mix and match.

SIP entities are all the network entities that are a part of the SIP System. Session Manager supports the following SIP entities which form part of the overall enterprise setup:

- Avaya Aura[™] Session Manager
- Avaya Aura[™] System Manager
- Private branch exchanges
- Avaya Aura[™] Communication Manager Feature Server
- Public switched telephone network service providers
- SIP-enabled adjuncts that work with PBXs to provide services
- Desksets
 - Avaya one-X desksets
 - 3rd Party desksets
- Avaya Aura[™] one-X Communicator
- Avaya G860 High Density Trunk Gateway R2 that work with non-SIP service provider networks
- Communication Manager with Branch Session Manager
- Branch Gateway
- Acme Packets Session Border Controller 2900/3800/4500

Session Manager is the software component for all enterprise SIP sessions. System Manager is the single, centralized management point of control. It runs on a separate server, and only one is required for the entire network, not per instance.

Related topics:

SIP entities on page 14
Secure Access Link on page 16
Survivable Core on page 16
Survivable Remote on page 17

SIP entities

Session Manager, Release 6.0, supports several PBXs, gateways, telephones, and other SIP entities and SIP-enabled adjuncts.

Communication Manager Feature servers

Communication Manager as a feature server provides Communication Manager features to SIP endpoints using the IP Multimedia Subsystem (IMS) half call model that allows full application sequencing.

The feature server only supports SIP endpoints that are registered to Avaya Aura[™] Session Manager. The Communication Manager server is connected to Session Manager via a SIP-ISC interface which uses an IMS-enabled SIP signaling group and associated SIP trunk group. Communication Manager is administered as a feature server by enabling the IMS-enabled? field on the signaling group form.

Communication Manager as a feature server has the following constraints:

- The dial plan for IMS users must route all PSTN calls back to Session Manager over the IMS trunk group. Routing of such calls directly to ISDN trunks is not supported.
- IPSI port networks are not supported.
- Traditional phones such as DCP, H.323, ISDN, and analog are not supported.

Communication Manager Evolution Server

Communication Manager as an evolution server provides Communication Manager features to both SIP and non-SIP endpoints. It uses the full call model with Communication Manager as the only supported application.

With an evolution server:

- H.323, digital, and analog endpoints register with Communication Manager
- SIP endpoints register with Session Manager (URE user)
- All endpoints receive service from Communication Manager

Communication Manager is connected to the Session Manager via a SIP—ISC interface. The Session Manager routes calls from and to SIP endpoints. The SIP endpoints can then communicate with all other endpoints that are connected to the Communication Manager.

PBXs

A PBX serves a particular business or office, providing the telephone features to the employees. Session Manager connects both enterprise-wide and small key systems generally used in branch offices. Session Manager connects to and acts as a system-wide dial plan for call processing applications such as:

- Avaya Aura[™] Communication Manager and later using direct SIP connections. Connections using SIP Enablement Services (SES) are not supported.
- SIP-enabled Cisco Unified CallManager.
- Nortel CS1000 SIP-enabled PBX.

SIP Gateways

SIP gateways work with the non-SIP service provider network. Session Manager works with SIP peer systems within the enterprise in the same way. These include non-Avaya SIP systems, SIP-enabled PBXs, and Communication Manager. Such SIP gateways are supported both for trunking services or line-side services. SIP gateways include trunk gateways, such as the Avaya G860 High Density Trunk Gateway R2.

SIP PSTN service providers

Service providers are treated as SIP peer network elements with which Session Manager maintains a trunking relationship. Foreign domain PBXs or SIP switching equipment are treated essentially the same as the SIP service providers, that is, as a SIP peer network element over a SIP trunk. Service providers supported include AT&T and Verizon.

SIP-enabled adjuncts

SIP-enabled adjuncts provide supplemental services to PBXs, such as voice mail and conferencing capabilities. Avaya products supported include Avaya Voice Portal 5.1 and 6.0, Avaya Modular Messaging Release 6.0, 5.2 and 5.1, Avaya Meeting Exchange 6.0 and 5.1.

SIP devices

SIP devices, particularly the Avaya 96XX handsets R2.6, may register to the Session Manager core. Session Manager can support up to 50,000 Avaya SIP devices. Session Manager provides SIP proxy, registrar, location services, and more to this initial set of devices.

Secure Access Link

Secure Access Link (SAL) enables monitoring the equipment and software. It is centralized and accessible through System Manager and has access to all the Session Manager equipment and software. SAL provides alarming and logging.

The major Session Manager components that are monitored include

- Server and all applicable components
- Operating system
- System Manager platform, including Routing
- · Session Manager solution.

Survivable Core

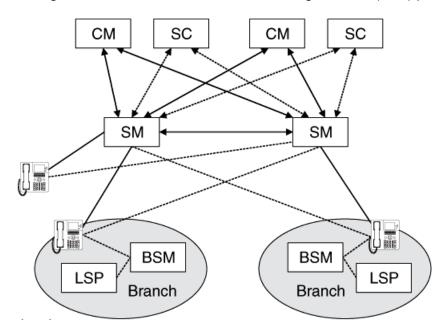
Survivable Core provides geo-redundant Feature Server redundancy and supports multiple Data Centers for a failed or unreachable main Communication Manager. The Survivable Core works as per following conditions:

- Session Manager monitors entity links to the Core Communication Managers via OPTIONs.
- Once Core Communication Manager goes down, the Session Manager starts sending SIP messages to the Survivable Core.

- When connectivity to the main Communication Manager is restored, Gateways reregister with the main Communication Manager automatically via Media Gateway Recovery Rules.
- When the main Communication Manager recovers, Session Manager again starts sending SIP messages to the Communication Manager instead of the Survivable Core.

Survivable Remote

Survivable Remote Sites include a local Branch Session Manager and local Communication Manager (either a Feature Server or an Evolution Server). SIP Phones simultaneously register to the main Session Manager, a backup main Session Manager and the Local Branch Session Manager. During a WAN outage, the phones failover to the Branch Survivable Session Manager and the local Communication Manager server (LSP) provides feature functionality.



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BSM	Branch Session Manager
СМ	Communication Manager (Feature Server and Evolution Server)
LSP	Local Survivable Processor
SC	Survivable Core

Survivable Remote provides the following features:

- Intelligently works around various individual elements and connectivity outages:
 - with Communication Manager at the core
 - with Session Manager at the core
 - WAN outages
- Active-Active geo-redundancy support
- SIP phones support for simultaneous registration to multiple controllers for fast failover and failback

High availability

Session Manager provides redundancy by supporting up to six Session Manager instances in an enterprise and are implemented in the same data center or in data centers that are separated geographically, even around the world. These instances need not exist on the same subnet.

Session Manager redundancy supports networks with round trip delays of less than 1 second.

Session Manager uses the active-active approach where two instances are simultaneously active, meaning any request goes to either instance and is served. Requests use either instance freely. This is important for distributing traffic across the network.

Active-active redundancy requires that the Session Manager instances be interconnected over an IP network with sufficient bandwidth and low enough latency to synchronize runtime data.

Configuring more than one Session Manager in a network means that:

- A failure of one of the Session Manager instances does not interrupt service. All calls that
 were stable before the failure remain stable with no discernible gaps in voice connectivity.
 New calls may be placed immediately after the failure with no more than 3 seconds of
 delay.
- A centralized enterprise-wide dial plan can be entered and managed using terminals logically connected to the server running System Manager.
- The centralized dial plan governs Avaya and third party PBXs (particularly Nortel and Cisco) and enables them to connect via SIP (either directly or via a SIP gateway) to one of the Session Manager instances.

Security

Session Manager is the SIP routing element between SIP entities. As the SIP router, all SIP sessions flow through Session Manager, allowing it to provide the following security capabilities to centralized SIP applications:

- TLS session connection termination for SIP trunks
- Network/transport firewall and denial of service (DoS) protection
- SIP DoS protection
- Access control to Avaya applications where only authenticated SIP elements are allowed access using System Manager Trust Management functionality
- · Whitelisting and blacklisting of SIP elements.

System Manager

Central management of Session Manager is handled through the Avaya Aura[™] System Manager application. System Manager delivers a set of shared, secure management services and a common console across multiple products. System Manager includes the following central management services:

- User Management: Allows for the administration of users and user groups.
- Communication System Management: Allows for the administration of individual and group stations and mailboxes.
- Routing: Allows for the administration of routing policies for all Session Manager instances within an enterprise.
- Alarm Management Service: Supports alarm monitoring, acknowledgement, configuration, clearing, and retiring.
- Logging Service: Receives log events formatted in the common log format.
- Enterprise Licensing Management Service

A central database that resides on the System Manager server stores all the System Manager central data, the Session Manager administration data, and the Central Data Distribution Service information. The Central Data Distribution Service detects changes to the System Manager central database and distribute these changes to the Session Manager instances. All communication between System Manager and Session Manager instances is done over secure links.

Licensing

Session Manager requires a concurrent user license. When Communication Manager 6.0 licenses are ordered through Avaya Solution Designer (ASD), these licenses can be included in a Session Manager license file on the Avaya Product Licensing and Delivery System Web site. If an enterprise uses application sequencing capabilities of Session Manager for users registered to Session Manager but do not have Communication Manager 6.0.1 licenses, additional licenses for these types of users may also be obtained from Avaya Solution Designer.

Application sequencing is also possible for some of the following endpoints that are not registered to Session Manager:

- DCP
- Analog
- H.323 endpoints on Communication Manager Evolution Server
- Users on third party PBXs where those PBXs are SIP-connected to Session Manager
- Dialed Number Identification Service (DNIS) or Direct Inward Dial (DID) numbers on incoming trunks to Session Manager

Related documents

Session Manager comes with a complete set of documents. The following list provides the title, document number, and a brief description of all of the documents in the documentation set.

- Avaya Aura[™] Session Manager Overview (03–603323) Provides descriptions of Session Manager and its components.
- Installing and Configuring Avaya Aura[™] Session Manager (03–603473) Provides information for installing Session Manager and the Avaya S8800 server, and provides initial administration for Session Manager.
- Administering Avaya Aura[™] Session Manager (03–603324) Provides information on administering Session Manager through System Manager.

- Maintaining and Troubleshooting Avaya Aura[™] Session Manager (03–603325) Provides information on maintaining and troubleshooting Session Manager, including logging and alarming.
- Security Design for Avaya Aura[™] Session Manager Provides information on making Session Manager secure on the network.

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