

CASE STUDY

ticketmaster

Ticketmaster Case Study

High Quality VoIP Connectivity and Professional Services for a Global Genesys Contact Center

<u>Ticketmaster</u> one of four business units within Live Nation Entertainment, is the global event ticketing leader and one of the world's top five eCommerce sites, with over 32 million monthly unique visitors. Offering global entertainment ticket sales, Ticketmaster services 20 different international markets with 14 contact centers around the globe.

Challenges

Experiencing growth in ticket sales and faced with aging, costly and disparate ACD systems, Ticketmaster embarked on a technology refresh initiative for their customer-facing contact centers with set goals to improve customer service and fan satisfaction, support global expansion, and reduce operating costs.

Elizabeth Gotto, senior vice president Global Contact Center Technology for Ticketmaster notes: "Ticketmaster's contact center was in need of a complete technology overhaul, eliminating the separate ACDs that were reaching endof-life, at capacity, expensive to maintain and becoming cumbersome to our delivering world-class service to our fans."

To achieve their goals, Ticketmaster developed a technology refresh plan that would ultimately have them implement one of the largest Genesys SIP contact centers, earning them the prestigious "Genesys Customer Innovation Award" in 2011:

- 1. Consolidate the contact center technology into a handful of redundant data-centers, starting with the US and Canada
- 2. Soon thereafter, implement similar data-centers in other geographies, creating a common global contact center platform
- 3. Leverage SIP to connect contact center agents, allowing for remote agent centers, a dramatic reduction in equipment costs and improve agent efficiency
- 4. Consolidate trunking facilities, leveraging SIP Trunking when practical to reduce last mile facilities and per-minute charges
- 5. Implement new multi-media personalization, improving the fan experience
- 6. Improve reporting and data collection, providing management significantly improved visibility to the operational status of the contact center.
- 7. Provide a better experience to fans complementing the Contact Center World Class Service program



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Challenges

However, accomplishing the above plan included a number of unique technical challenges, requiring careful planning with a sound three-step migration strategy:

- 1) Interfacing the existing TDM toll-free trunk lines to the new SIP-based Genesys contact center, allowing the existing voice carrier contracts to remain in a "like-for-like" configuration.
- 2) As soon as the new Genesys contact center system transition is complete and stable, initiate a migration to SIP trunking and thereby, reduce costs by eliminating redundant TDM last-mile facilities.

"AudioCodes has been one of the most reliable and stable components in the infrastructure", commented Elizabeth.

Elizabeth Gotto, Senior Vice President Global Contact Center Technology

3) The final hurdle is to bring on-line new all-IP contact center sites in EMEA and other regions, integrating them into a single global network, requiring implementation and validation of new global IP network connectivity using a complex mesh network topology.

"A key requirement for Ticketmaster was to find a best-of-breed supplier that could deliver a gateway solution that could be used in all our global data centers," commented Gotto. "While at the same time, we wanted those same gateways to function as an Enterprise Session Border Controller (E-SBC) in the future."

Solution

To facilitate their global evolution to SIP, Ticketmaster turned to AudioCodes.

During the first "Like-for-Like" phase, Ticketmaster selected the AudioCodes Voice Gateway to provide the interface between the TDM trunk lines and Genesys SIP Contact Center. With its three high-density DS3 physical interfaces and optional High Availability redundancy, the Mediant 3000 proved to be a reliable and scalable foundation for Ticketmaster as they began their transition to SIP.

Once the US data center sites were complete, the second phase of Ticketmaster's transition to SIP was initiated, integrating SIP Trunking using E-SBCs as a point of demarcation, solving interoperability with the SIP Trunking providers and improving security. Since Ticketmaster already owned a number of AudioCodes Mediant 3000 Media Gateways, this was an easy upgrade, facilitated by purchasing optional E-SBC software licenses for those already installed gateways. With E-SBC functionality installed, Ticketmaster was able to move the North American data centers to Verizon IP Toll-Free trunking, reducing their facilities charges and adding call routing flexibility.

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When the time came to expand the Genesys SIP contact center deployment beyond North America, Ticketmaster needed to ensure service levels and voice quality would be maintained across the expanded network. To accomplish this, Ticketmaster turned to the Professional Services team at AudioCodes to perform a global Network Readiness Assessment, the goal of which was to verify the global network performance and avoid any unforeseen issues that would cause voice quality issues that might negatively affect the fan experience.

Results

The goals of significantly improving agent efficiency, reducing operational costs and improving call data tracking have largely been realized by Ticketmaster - a side effect of their implementation of Genesys SIP contact center software operating with voice network infrastructure from AudioCodes. Almost 100% of Ticketmaster's call traffic now enters their North American contact center via SIP Trunks, utilizing AudioCodes Mediant 3000 E-SBCs to provide interoperability and security.

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When discussing the results of the global Network Readiness Assessment performed by AudioCodes Professional Services, Gotto noted that "the Network Readiness Assessment brought to light some network issues that we had no idea existed - had we not done the assessment before going live, we surely would have suffered down-time." As a result, Ticketmaster was able to point out some specific issues to the network team and providers that linked their North American and Amsterdam sites, thereby eliminating a significant voice quality issue in advance of going live.

And in a closing comment, Gotto noted that "Genesys is highly dependent on the network it rides on. Ensuring your network is ready to support a SIP deployment is imperative." proving that Ticketmaster's investment in AudioCodes gateways, SBCs and Network Readiness Assessment was a wise one.

For more information on AudioCodes and Genesys SIP solutions, visit: www.audiocodes.com/genesys

Mediant 3000 E-SBC

AudioCodes' Mediant 3000 Enterprise Session Border Controller (E-SBC) is a member of AudioCodes family of Enterprise Session Border Controllers, enabling connectivity and security between enterprises and service provider VoIP networks.



The Mediant 3000 E-SBC provides protocol mediation for allowing the seamless connection of any PBX and/or IP-PBX to any service provider, perimeter defense as a way of protecting companies from malicious VoIP attacks, and service assurance for service quality and manageability. Designed as a cost-effective appliance, the Mediant 3000 E-SBC is based on field-proven VoIP and network services with a native host processor and integrated quality of service mechanisms. The native implementation of SBC functions on the Mediant 3000 provides a host of additional capabilities, such as VoIP mediation, PSTN Access, survivability, and the ability to host third party value-added service applications. This enables enterprises to utilize the advantages of converged networks and eliminate the need for standalone appliances.

About AudioCodes

AudioCodes Ltd. (NasdagGS: AUDC) designs, develops and sells advanced Voice over IP (VoIP) and converged VoIP and Data networking products and applications to Service Providers and Enterprises. AudioCodes is a VoIP technology market leader focused on converged VoIP & data communications and its products are deployed globally in Broadband, Mobile, Cable, and Enterprise networks. The company provides a range of innovative cost-effective products including Media Gateways, Multi-Service Session Business Routers, Border Controllers (SBC), Residential Gateways, IP Phones, Media Servers and Value Added Applications. AudioCodes' underlying technology, VolPerfectHD™, relies on AudioCodes' leadership in DSP, voice coding and voice processing technologies. AudioCodes High Definition (HD) VoIP technologies and products provide enhanced intelligibility and a better end user communication experience in Voice communications.

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