

Architectural and Engineering Specifications

The solution shall be a complete audio solution for conference rooms and conference spaces. All components in the solution shall be Dante enabled and PoE or PoE+ powered, requiring only one CAT5e or Cat6 cable for both power and data/audio transmission to the devices. The solution shall include one or two ceiling microphones, two wall speakers, an audio processor, and a network switch all from one manufacturer. The ceiling microphone shall deploy dynamic beam-forming technology. It shall support multi-beams to be able to capture several areas of the room simultaneously. The microphone beams shall be following active speakers when moving in the room. When a new person starts speaking either a pre-existing beam shall capture that person immediately, or a new beam shall form within milliseconds on that location. Different mixing options to handle the active beams shall be offered, including gain-sharing options and all-mix. The ceiling microphone shall offer three options for installation. These options shall include methods to mount it flush below a ceiling tile, to mount it hanging from a ceiling using wires, and to mount it using a VESA mount. The ceiling microphone shall include audio post-processing. This shall include Adaptive Acoustic Echo Cancellation, Noise Reduction, Human Voice Activity Detection, Automatic Gain Control, Automatic mixing of audio signals, dereverberation, and others. The microphone shall offer different kinds of mixing technology, including "matrix mixing" and "gain-share mixing". The ceiling microphone shall provide an audio signal that was only post-processed using linear algorithms allowing for technology like speech recognition to use the signal without loss of performance. The wall speakers shall be line-array speakers with 16 speaker elements each. Two speakers shall be included in the solution. The line array wall speakers shall create a directionality to the audio signal, resulting in reduced signal loss over distance. The line array wall speakers shall be specifically designed for communication and conferencing applications, with emphasis on the typical frequency spectrum of human voices. The room audio processor shall support a means to detect the separate solution devices like ceiling microphone and wall speakers and auto-configure the audio flow between these devices. As part of that process it shall configure the devices to communicate correctly with each other. It also shall provide a mode to automatically measure the acoustic in the room and tune the devices to the specific room acoustic. The room audio processor shall integrate with standard call managers using SIP* telephony. It furthermore shall support Bluetooth to connect to cell phones. USB audio shall be supported for integration with Unified Communications clients. The room audio processor shall support analog audio for connections with video appliances/codecs. The room audio processor shall support two analog microphone inputs for presenters in the room. These inputs shall allow voice lift for audio on these microphones, but at the same time mix the audio into the outgoing stream on USB, BT, analog audio, and SIP*. Feed-back suppression shall be supported on these microphone inputs. A network connection into the room audio processor shall allow for remote management of the overall solution from a corporate network. The network switch shall be approved and pre-configured to be used in Dante networks. It shall support PoE+ support on eight connections, up to 240 Watts. The solution shall provide connectivity for third party control systems, allowing to manage system, microphone, and audio behavior from the room control system. The system shall provide information on beam direction allowing room control systems to use that information for example to steer motorized PTZ cameras. It shall be remotely accessible for monitoring and managing. The solution shall be able to send proactive information on its state and possible problems to a management system. The Yamaha ADECIA ceiling solution is specified.

* SIP functionality is not supported in firmware versions 3.0 and later.