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DIRECTIONAL ACTIVITY MASK DETECTOR FOR A VEHICLE

Abstract

A method for a directional activity mask detector for a vehicle includes generating a blocking matrix based on pre-recorded signals from a target zone, receiving, at a voice activity detector, audio frames from a microphone array, and applying the blocking matrix to one or more zones within a vehicle. The method also includes detecting signals from unblocked zones of the vehicle, determining an activity of a target signal based on the detected signals from the unblocked zones, and estimating, by a beamformer, a relative transfer function (RTF) vector based on the received audio frames and the determined activity of the target signal.

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Background/Summary

INTRODUCTION

[0001] The information provided in this section is for the purpose of generally presenting the context of the disclosure. Work of the presently named inventors, to the extent it is described in this section, as well as aspects of the description that may not otherwise qualify as prior art at the time of filing, are neither expressly nor impliedly admitted as prior art against the present disclosure.

[0002] Speech enhancement is required for both virtual assistant and hands-free applications. The present disclosure relates generally to a directional activity mask detector based on algebraic blocking matrix for detecting a target speaker in a vehicle acoustic environment and ignoring directional interference.

[0003] A beamformer is commonly used for speech enhancement of a target signal using a microphone array according to its direction. The target direction is defined by a steering vector, estimated by Relative transfer function tracking (RTFT) according to a target speech activity tracking. Often, the activity of a target signal is unknown or estimated. If the RTFT is updated with the wrong direction it may result in self-cancellation. Further, imprecise target activity detection may contaminate the direction tracking and may result in leakage of interference into a target output signal. Thus, there is a need to improve the activity detection of the target speaker when analyzing audio signals within a vehicle.

[0004] The proposed mask is applied on the received audio signal to indicate the RTFT adaptation according to active time frequency bins related to the correct direction of the target speaker.

SUMMARY

[0005] One aspect of the disclosure provides a computer-implemented method for a directional activity mask detector for a vehicle that, when executed by data processing hardware, causes the data processing hardware to perform operations that include generating a blocking matrix based on pre-recorded signals from a target zone, receiving, at a voice activity detector, audio frames from a microphone array, and applying the blocking matrix to one or more zones within a vehicle. The operations also include detecting signals from unblocked zones of the vehicle, determining an activity of a target signal based on the detected signals from the unblocked zones, and estimating, by a beamformer, a relative transfer function (RTF) vector based on the received audio frames and the determined activity of the target signal.

[0006] Implementations of the disclosure may include one or more of the following optional features. In some examples, the operations include defining a blocking area of the blocking matrix. Optionally, the operations include tracking a noise floor with an energy detector. In some implementations, the operations include generating an energy threshold by Monte Carlo simulation. In these implementations, the operations may also include identifying an optimal energy threshold based on the energy threshold generated by the Monte Carlo simulation. Here, the operations may further include tailoring the identified optimal energy threshold for an audio task. Optionally, generating the blocking matrix include recording clean signals from each zone. In some implementations, the operations further include enhancing the received audio frames by transforming the audio frames into a short-time Fourier transform (STFT) domain.

[0007] Another aspect of the disclosure provides a system for a directional activity mask detector for a vehicle that includes data processing hardware and memory hardware in communication with the data processing hardware. The memory hardware stores instructions that when executed by the data processing hardware cause the data processing hardware to perform operations that include generating a blocking matrix based on pre-recorded signals from a target zone, receiving, at a voice activity detector, audio frames from a microphone array, and applying the blocking matrix to one or more zones within a vehicle. The operations also include detecting signals from unblocked zones of the vehicle, determining an activity of a target signal based on the detected signals from the unblocked zones, and estimating, by a beamformer, an RTF vector based on the received audio frames and the determined activity of the target signal.

[0008] This aspect may include one or more of the following optional features. In some examples, the operations include defining a blocking area of the blocking matrix. Optionally, the operations include tracking a noise floor with an energy detector. In some implementations, the operations include generating an energy threshold by Monte Carlo simulation. In these implementations, the operations may also include identifying an optimal energy threshold based on the energy threshold generated by the Monte Carlo simulation. Here, the operations may further include tailoring the identified optimal energy threshold for an audio task. Optionally, generating the blocking matrix include recording clean signals from each zone. In some implementations, the operations further include enhancing the received audio frames by transforming the audio frames into a short-time Fourier transform (STFT) domain.

[0009] Another aspect of the disclosure provides a system for a directional activity mask detector for a vehicle that includes data processing hardware and memory hardware in communication with the data processing hardware. The memory hardware stores instructions that when executed by the data processing hardware cause the data processing hardware to perform operations that include generating a blocking matrix based on a steering vector. Here, the blocking matrix includes a mask. The operations also include receiving, at a voice activity detector, audio frames from a microphone array, and applying the blocking matrix to one or more zones within the vehicle. The operations further include detecting signals from unblocked zones of the vehicle and determining an activity of a target signal based on the detected signals.

[0010] This aspect may include one or more of the following optional features. In some examples, the operations further include calculating a ratio of energy changes between a reference microphone signal and a maximum value of outputs of the blocking matrix. In these examples, the operations may further include generating the mask of the blocking matrix based on the ratio of energy changes. Here, the operations may also include identifying active bins of the mask and updating a relative transfer function (RTF) based on the identified active bins.

Description

BRIEF DESCRIPTION OF THE DRAWINGS

[0011] The drawings described herein are for illustrative purposes only of selected configurations and are not intended to limit the scope of the present disclosure.

[0012] FIG. 1 is a schematic of a vehicle equipped with a speech enhancement system according to the present disclosure;

[0013] FIG. 2 is a partial perspective view of an interior cabin of a vehicle with passengers emitting audio signals captured by a speech enhancement system according to the present disclosure;

[0014] FIG. 3 is a functional block diagram of a speech enhancement system according to the present disclosure;

[0015] FIG. 4 is a schematic of an interior of a vehicle according to the present disclosure with defined zones;

[0016] FIG. 5 is an example schematic of a speech enhancement system according to the present disclosure;

[0017] FIG. 6 is another example schematic of a speech enhancement system according to the present disclosure;

[0018] FIG. 7 is an example flow diagram for a speech enhancement system according to the present disclosure; and

[0019] FIG. 8 is a continued example flow diagram for the speech enhancement system of FIG. 7.

[0020] Corresponding reference numerals indicate corresponding parts throughout the drawings.

DETAILED DESCRIPTION

[0021] Example configurations will now be described more fully with reference to the

accompanying drawings. Example configurations are provided so that this disclosure will be thorough, and will fully convey the scope of the disclosure to those of ordinary skill in the art. Specific details are set forth such as examples of specific components, devices, and methods, to provide a thorough understanding of configurations of the present disclosure. It will be apparent to those of ordinary skill in the art that specific details need not be employed, that example configurations may be embodied in many different forms, and that the specific details and the example configurations should not be construed to limit the scope of the disclosure.

[0022] The terminology used herein is for the purpose of describing particular exemplary configurations only and is not intended to be limiting. As used herein, the singular articles “a,” “an,” and “the” may be intended to include the plural forms as well, unless the context clearly indicates otherwise. The terms “comprises,” “comprising,” “including,” and “having,” are inclusive and therefore specify the presence of features, steps, operations, elements, and/or components, but do not preclude the presence or addition of one or more other features, steps, operations, elements, components, and/or groups thereof. The method steps, processes, and operations described herein are not to be construed as necessarily requiring their performance in the particular order discussed or illustrated, unless specifically identified as an order of performance. Additional or alternative steps may be employed.

[0023] When an element or layer is referred to as being “on,” “engaged to,” “connected to,” “attached to,” or “coupled to” another element or layer, it may be directly on, engaged, connected, attached, or coupled to the other element or layer, or intervening elements or layers may be present. In contrast, when an element is referred to as being “directly on,” “directly engaged to,” “directly connected to,” “directly attached to,” or “directly coupled to” another element or layer, there may be no intervening elements or layers present. Other words used to describe the relationship between elements should be interpreted in a like fashion (e.g., “between” versus “directly between,” “adjacent” versus “directly adjacent,” etc.). As used herein, the term “and/or” includes any and all combinations of one or more of the associated listed items.

[0024] The terms “first,” “second,” “third,” etc. may be used herein to describe various elements, components, regions, layers and/or sections. These elements, components, regions, layers and/or sections should not be limited by these terms. These terms may be only used to distinguish one element, component, region, layer or section from another region, layer or section. Terms such as “first,” “second,” and other numerical terms do not imply a sequence or order unless clearly indicated by the context. Thus, a first element, component, region, layer or section discussed below could be termed a second element, component, region, layer or section without departing from the teachings of the example configurations.

[0025] In this application, including the definitions below, the term “module” may be replaced with the term “circuit.” The term “module” may refer to, be part of, or include an Application Specific Integrated Circuit (ASIC); a digital, analog, or mixed analog/digital discrete circuit; a digital, analog, or mixed analog/digital integrated circuit; a combinational logic circuit; a field programmable gate array (FPGA); a processor (shared, dedicated, or group) that executes code; memory (shared, dedicated, or group) that stores code executed by a processor; other suitable hardware components that provide the described functionality; or a combination of some or all of the above, such as in a system-on-chip.

[0026] The term “code,” as used above, may include software, firmware, and/or microcode, and may refer to programs, routines, functions, classes, and/or objects. The term “shared processor” encompasses a single processor that executes some or all code from multiple modules. The term “group processor” encompasses a processor that, in combination with additional processors, executes some or all code from one or more modules. The term “shared memory” encompasses a single memory that stores some or all code from multiple modules. The term “group memory” encompasses a memory that, in combination with additional memories, stores some or all code from one or more modules. The term “memory” may be a subset of the term “computer-readable

medium.” The term “computer-readable medium” does not encompass transitory electrical and electromagnetic signals propagating through a medium, and may therefore be considered tangible and non-transitory memory. Non-limiting examples of a non-transitory memory include a tangible computer readable medium including a nonvolatile memory, magnetic storage, and optical storage. [0027] The apparatuses and methods described in this application may be partially or fully implemented by one or more computer programs executed by one or more processors. The computer programs include processor-executable instructions that are stored on at least one non-transitory tangible computer readable medium. The computer programs may also include and/or rely on stored data.

[0028] A software application (i.e., a software resource) may refer to computer software that causes a computing device to perform a task. In some examples, a software application may be referred to as an “application,” an “app,” or a “program.” Example applications include, but are not limited to, system diagnostic applications, system management applications, system maintenance applications, word processing applications, spreadsheet applications, messaging applications, media streaming applications, social networking applications, and gaming applications.

[0029] The non-transitory memory may be physical devices used to store programs (e.g., sequences of instructions) or data (e.g., program state information) on a temporary or permanent basis for use by a computing device. The non-transitory memory may be volatile and/or non-volatile addressable semiconductor memory. Examples of non-volatile memory include, but are not limited to, flash memory and read-only memory (ROM)/programmable read-only memory (PROM)/erasable programmable read-only memory (EPROM)/electronically erasable programmable read-only memory (EEPROM) (e.g., typically used for firmware, such as boot programs). Examples of volatile memory include, but are not limited to, random access memory (RAM), dynamic random access memory (DRAM), static random access memory (SRAM), phase change memory (PCM) as well as disks or tapes.

[0030] These computer programs (also known as programs, software, software applications or code) include machine instructions for a programmable processor, and can be implemented in a high-level procedural and/or object-oriented programming language, and/or in assembly/machine language. As used herein, the terms “machine-readable medium” and “computer-readable medium” refer to any computer program product, non-transitory computer readable medium, apparatus and/or device (e.g., magnetic discs, optical disks, memory, Programmable Logic Devices (PLDs)) used to provide machine instructions and/or data to a programmable processor, including a machine-readable medium that receives machine instructions as a machine-readable signal. The term “machine-readable signal” refers to any signal used to provide machine instructions and/or data to a programmable processor.

[0031] Various implementations of the systems and techniques described herein can be realized in digital electronic and/or optical circuitry, integrated circuitry, specially designed ASICs (application specific integrated circuits), computer hardware, firmware, software, and/or combinations thereof. These various implementations can include implementation in one or more computer programs that are executable and/or interpretable on a programmable system including at least one programmable processor, which may be special or general purpose, coupled to receive data and instructions from, and to transmit data and instructions to, a storage system, at least one input device, and at least one output device.

[0032] The processes and logic flows described in this specification can be performed by one or more programmable processors, also referred to as data processing hardware, executing one or more computer programs to perform functions by operating on input data and generating output. The processes and logic flows can also be performed by special purpose logic circuitry, e.g., an FPGA (field programmable gate array) or an ASIC (application specific integrated circuit). Processors suitable for the execution of a computer program include, by way of example, both general and special purpose microprocessors, and any one or more processors of any kind of digital

computer. Generally, a processor will receive instructions and data from a read only memory or a random access memory or both. The essential elements of a computer are a processor for performing instructions and one or more memory devices for storing instructions and data. Generally, a computer will also include, or be operatively coupled to receive data from or transfer data to, or both, one or more mass storage devices for storing data, e.g., magnetic, magneto optical disks, or optical disks. However, a computer need not have such devices. Computer readable media suitable for storing computer program instructions and data include all forms of non-volatile memory, media and memory devices, including by way of example semiconductor memory devices, e.g., EPROM, EEPROM, and flash memory devices; magnetic disks, e.g., internal hard disks or removable disks; magneto optical disks; and CD ROM and DVD-ROM disks. The processor and the memory can be supplemented by, or incorporated in, special purpose logic circuitry.

[0033] To provide for interaction with a user, one or more aspects of the disclosure can be implemented on a computer having a display device, e.g., a CRT (cathode ray tube), LCD (liquid crystal display) monitor, or touch screen for displaying information to the user and optionally a keyboard and a pointing device, e.g., a mouse or a trackball, by which the user can provide input to the computer. Other kinds of devices can be used to provide interaction with a user as well; for example, feedback provided to the user can be any form of sensory feedback, e.g., visual feedback, auditory feedback, or tactile feedback; and input from the user can be received in any form, including acoustic, speech, or tactile input. In addition, a computer can interact with a user by sending documents to and receiving documents from a device that is used by the user; for example, by sending web pages to a web browser on a user's client device in response to requests received from the web browser.

[0034] Referring now to the Figures, a speech enhancement system **10** is described with respect to a vehicle **100** equipped with a microphone array **102** configured to receive audio signals **104**. With reference to FIG. 2, an interior cabin **106** of the vehicle **100** is illustrated with a driver and a passenger. The driver is illustrated as emitting an audio signal **104**, which is captured by the microphone array **102**. However, the passenger or passengers may also emit the audio signals **104**. Further, the audio signals **104** may include environmental noise. The microphone array **102** includes a reference microphone **102a**, which may be selected during a tuning process, described herein. It is contemplated that the speech enhancement system **10** may alter which microphone of the microphone array **102** may serve as the reference microphone **102a**. Thus, the reference microphone **102a** is selected according to the configuration of the microphone array **102** during a tuning process, as described in more detail below.

[0035] The vehicle **100** may be divided into a plurality of zones Z.sub.1-Z.sub.n. For example, FIG. 4 illustrates the vehicle **100** as being divided into four zones Z.sub.1-Z.sub.4. It is contemplated that the speech enhancement system **10** may be configured to divide the vehicle **100** into fewer than four zones or greater than four zones depending on the size of the vehicle **100** and analysis performed by the speech enhancement system **10**. The speech enhancement system **10** includes an electronic control unit (ECU) **12** of the vehicle **100** that is configured with a directional activity application **14**. The directional activity application **14** is executed by data processing hardware **16** of the ECU **12**. The ECU **12** may also include memory hardware **18** in communication with the data processing hardware **16**. The memory hardware **18** stores operative functions of the directional activity application **14**, described in more detail below.

[0036] With further reference to FIGS. 1-4, the directional activity application **14** includes a voice activity detector **20** that is utilized in generating a directional mask **22** in combination with an energy detector **24**. The energy detector **24** is configured with an energy threshold **26** and a noise floor **28**. The energy threshold **26** and the noise floor **28** may vary depending on the desired output from the speech enhancement system **10**. The audio signals **104** are received by the ECU **12** as audio frames **110**. The audio frames **110** are in a time domain when received by the microphone

array **102** and are converted into a frequency domain via a short-time Fourier transform (STFT) domain **30**. The conversion into the STFT domain **30** may be executed by the data processing hardware **16** of the ECU **12**. The directional activity application **14** receives the audio signals **104** as the converted frames **110** in real time. The memory hardware **18** may store an acoustic transfer function **36**, a delay and sum model **38**, and a blocking matrix **40**, described in more detail below. [0037] The directional activity application **14** utilizes the blocking matrix **40** in combination with the voice activity detector **20** and the energy detector **24** to generate the directional mask **22**. Thus, the directional mask **22** may be based on the blocking matrix **40**. For example, the blocking matrix **40** is designed to block a desired direction (e.g., a first zone Z.sub.1 with the driver) and capture the remaining audio signals **104** from the remaining zones Z.sub.2-Z.sub.4 and compare the audio signals **104** to the reference microphone **102a**. The audio signals **104** from the desired direction may be referred to herein as a target signal **104a** or target signals **104a**, and the remaining audio signals **104** may be referred to as an interference signal **104b** or interference signals **104b**.

[0038] The blocking matrix **40** is generated based on pre-recorded signals from an identified target zone. For example, the target signal(s) **104a** are recorded during an offline tuning process and may be utilized to develop the blocking matrix **40**. Once the blocking matrix **40** is developed, the blocking matrix **40** may be stored in the memory hardware **18** for selective use by the directional activity application **14**. The directional activity application **14** may project the frames **110** onto the generated blocking matrix **40**.

[0039] Referring to FIGS. 2-6, the ECU **12** receives the audio signals **104** from the microphone array **102** and may execute a beamformer **50** to enhance the target signal **104a**. The beamformer **50** utilizes a relative transfer function (RTF) **52**, which is determined by RTF tracking **54** to identify a steering or RTF vector **56** defined between the microphone array **102** and the target speaker **200** (e.g., the driver) to enhance the received target signals **104a**. For example, the ECU **12** may determine the steering vector **56** based on the received signals **104**.

[0040] In some examples, the voice activity detector **20** may be utilized to detect the target signal **104a** based on energy, and the directional activity application **14** may supply the directional mask **22** to the RTFT **54** to estimate the RTF **52** on the target signal **104a**. For example, the voice activity detector **20** receives frames **110** including the target signal **104a** from the microphone array **102**. The directional mask **22** may identify, from the audio signals **104**, the target signal **104a** according to the direction of each of the audio signals. Thus, the directional mask **22** provides a time frequency mask of the target signal **104a**.

[0041] Each audio signal **104** includes a frame **110** that is converted into the time frequency domain to determine which frequencies **202** of the audio signals **104** are dominated by the target speaker **200**. Differentiating the frames **110** assists in distinguishing the target signals **104a** from the interference signals **104b**, which may include other speakers and/or noise within the vehicle **100**. The directional activity application **14** distinguishes the frames **110** to determine the directivity of the audio signals **104** and identify the target signal(s) **104a**. For example, the directional mask **22** checks the directivity for a frequency of each frequency in each frame **110** to determine whether the directivity is comparable to the target signal(s) **104a**. Once the directional mask **22** has determined the directivity, the directional activity application **14** may execute the RTFT **34** for the audio signals **104**.

[0042] As described above, the vehicle **100** may be divided into a plurality of zones Z.sub.1-Z.sub.4, of which a single target zone is identified. The directional activity application **14** determines which, if any, zones Z.sub.1-Z.sub.4 are active and, within the active zones Z.sub.1-Z.sub.4, which frequencies **202**, if any, of the audio signals **104** are active. The directional activity application **14** makes this determination per frequency **202** and distinguishes between active and inactive audio signals **104**. For example, when a person is speaking within a target zone Z.sub.1-Z.sub.n there may be other speakers within the zone Z.sub.1-Z.sub.n that may register as inactive frequencies for the directional mask of that target zone Z.sub.1-Z.sub.n. The directional mask **22** is

utilized to create a timestamp of the frame **110** where there is activity to determine which audio signals **104** are active and which are not.

[0043] In some examples, some of the audio signals **104** may be from a driver, in a first zone Z.sub.1 and some of the audio signals **104** may be from a passenger, in a second zone Z.sub.2. The directional activity application **14** is configured to mask, via the directional mask **22**, and separate the two sets of audio signals **104** to identify which frequencies **202** of the audio signals **104** are the target signal(s) **104a**. The directional activity application **14** is configured to identify the directivity of the target signal(s) **104a** and mask, or cover, the target position to check the energy to determine the coverage of the target position. If the directional activity application **14** detects signals **104** after covering or masking the target position, then the directional activity application **14** may determine that the detected, remaining signals **104** are interference signals **104b**.

[0044] With further reference to FIGS. 2-6, the blocking matrix **40** is utilized to block the target zones Z.sub.1-Z.sub.n, which may be utilized in building the beamformer **50**. For example, the blocking matrix **40** is configured to block what is deemed to be the target zone Z.sub.1-Z.sub.n, and the directional activity application **14** checks the output to determine whether signals **104** are still getting past the blocking matrix **40**. In some examples, the blocking matrix **40** may be applied to one or more zones Z.sub.1-Z.sub.n within the vehicle **100**, and the directional activity application **14** may detect signals **104** from the unblocked zones to determine a directivity of the target signal **104a** based on the detected signals **104** from the unblocked zones.

[0045] The results from the directional activity application **14** may be utilized in building and refining the beamformer **50**. The blocking matrix **40** is developed, or built, to block the entire zone Z.sub.1-Z.sub.n of the target speaker **200**. The target speaker **200** is predefined depending on the task and may be positioned in different zones Z.sub.1-Z.sub.n depending on the scenario during a tuning process of the system **10**. For example, the directional activity application **14** may be dynamically adapted to assess the location of the target speaker **200** based on the activity of the target signals **104a**. By assessing the outputs of the blocking matrix **40**, in addition to the determined target position, the directional activity application **14** may more easily search for the target signal **104a**.

[0046] In some examples, a single reference microphone **102a** of the microphone array **102** is selected, and the energy detector **24** is applied to the reference microphone **102a**. The energy detector **24** is configured to avoid the noise floor **28**, such that the directionality is assessed when there is energy (e.g., a person speaking). For example, the directional activity application **14** may be configured to track the noise floor **28** within the energy detector **24**. The noise floor **28** is configured to mitigate capturing non-energy based signals or environmental noise, such as road noise. Thus, the directional activity application **14** is configured to ignore environmental noise. The directional activity application **14** further assumes that there is a speaker, such that if there is energy within a zone Z.sub.1-Z.sub.n and the zone Z.sub.1-Z.sub.n is covered, then a change is detected. For example, the directional activity application **14** is searching for the change as a result of blocking or otherwise masking a zone Z.sub.1-Z.sub.n.

[0047] For example, if a target speaker **200** is active (e.g., energy detected) and a change is detected in response to covering the zone Z.sub.1-Z.sub.n, then the target speaker **200** is speaking within the covered zone Z.sub.1-Z.sub.n regardless of position within the zone Z.sub.1-Z.sub.n. Thus, the activity of the target signal **104a** may be detected regardless of positioning of the target within zone Z.sub.1-Z.sub.n.

[0048] During enhancement of the target signal **104a**, the directional activity application **14** determines the energy per frequency compared to a previous frame **110** of the blocked zone Z.sub.1-Z.sub.n. If there is no change of the energy per frequency at the compared frames **110**, then the directional activity application **14** may determine that there are no searchable signals **104**. As noted above, the directional activity application **14** will utilize a reference microphone **102a** to search for the target signals **104a** by comparing the received signals **104** from the reference

microphone **102a** to the blocking matrix **40** output. The reference microphone **102a** is typically the closest microphone of the microphone array **102** to the target zone Z.sub.1-Z.sub.n. For example, the directional activity application **14** may search for an active zone Z.sub.1-Z.sub.n that contains several positions within the active zone Z.sub.1-Z.sub.n. Thus, the directional activity application **14** may select as the reference microphone **102a** the microphone closest to a target zone Z.sub.1-Z.sub.n of the active zone Z.sub.1-Z.sub.n. The reference microphone **102a** is, thus, configured to be a predefined feature of the directional activity application **14**, such that the reference microphone **102a** can change depending on the zone Z.sub.1-Z.sub.n relative to the orientation of the speaker. The reference microphone **102a** is predefined during the tuning process based on the target zone Z.sub.1-Z.sub.n, but may be altered during the tuning process to utilize a different microphone of the microphone array **102** as the reference microphone **102a**.

[0049] Referring still to FIGS. 2-6, the directional activity application **14** utilizes the blocking matrix **40** to block the target signal **104a**, and the energy detector **24** will detect a lowered energy. For example, the output of the blocking matrix **40** has N output channels, where the value of N is described in more detail below. The directional activity application **14** takes the maximum value of the N output channels and applies the energy detector **24** on the result. The outputs of the blocking matrix **40** may provide an indication as to whether a zone Z.sub.1-Z.sub.n is active or inactive. If the outputs of the blocking matrix **40** include the target signal **104a**, then the target signal **104a** may be blocked by all of the blocking matrix **40** outputs. Thus, the maximum value of the blocking matrix channels **40** will not contain the target signal **104a**. The directional activity application **14** calculates a ratio of energy changes between the received audio signals **104** and the received audio signals **104** after applying the blocking matrix **40**. For example, the ratio of energy changes may be calculated between the reference microphone **102a** signal and a maximum value of outputs of the blocking matrix **40**. Further, the directional mask **22** may be generated based on the ratio of energy changes. When a target speaker **200** is active, the ratio will be higher than when there is no activity within a blocked zone Z.sub.1-Z.sub.n.

[0050] FIG. 6 illustrates an example scenario where a target signal **104a** is received by the reference microphone **102a** and, on a lower branch, blocked by the blocking matrix **40**. The target signal **104a**, thus, passes through an upper branch of the directional activity application **14** through the reference microphone **102a** and is processed by the energy detector **24** to generate a delta value ($\Delta_{\text{sub.in}}$). The delta value generated along the upper branch is taken as a ratio with a delta value ($\Delta_{\text{sub.BM}}$) of the target signal **104a** generated along the lower branch. The delta value from the lower branch of the directional activity application **14** is generated based on the energy detector **24** evaluating the signal **104a** blocked by the blocking matrix. The energy detector **24** is applied on the maximum value of the P output channels of the blocking matrix. P is a parameter of the blocking matrix **40** that is less than M the number of microphones of the microphone array **102**. When building the blocking matrix **40**, the system **10** records several positions within the zones Z.sub.1-Z.sub.n and uses the acoustic transfer function **36** to build the blocking matrix **40**. Thus, the system **10** takes positions within the zones Z.sub.1-Z.sub.n to build a zone direction.

[0051] In establishing the blocking matrix, the directional activity application **14** is tuned from each zone Z.sub.1-Z.sub.n defined within the vehicle **100**. The recordings from each zone Z.sub.1-Z.sub.n are executed as clean recordings without noise to estimate an acoustic system of the vehicle **100**. The representation of the target speaker **200** seating area may be estimated using the following exemplary equation (a):

$$[00001] \ x_l(k, n) = H_l(k)s(k, n) \quad (a)$$

where $x_{\text{sub.l}}(k, n)$ is the audio signals **104** and $s(k, n)$ is a known target signal **104a**, before propagating in the vehicle **100**, and $H_{\text{sub.l}}(k)$ is the acoustic transfer function relating the microphones and the l-th seating position of the speaker. $H_{\text{sub.l}}(k)$ may be estimated using various models. For example, $H_{\text{sub.l}}(k)$ may be calculated based on a mathematical model **36** of the

various zones Z.sub.1-Z.sub.n and may be trained based on the signals s(k, n) and x.sub.l(k, n). The estimation may be dependent upon the construction and interior layout of the vehicle **100**, so the approach of building the blocking matrix **40** may be geometric and tailored specifically to the vehicle **100**.

[0052] In other examples, H.sub.l(k) may be modeled using a delay and sum model **38**, which assumes that H.sub.l(k) only contains the delay from the direct path of the ATF **36**, for each microphone in the microphone array **102**, using the following exemplary equation (b):

$$[00002] \ H_l(k) = [e^{j\frac{2\pi k}{K}\tau_{1,l}}, e^{j\frac{2\pi k}{K}\tau_{2,l}}, \dots, e^{j\frac{2\pi k}{K}\tau_{M,l}}]^T \quad (b)$$

Where $\tau_{m,l}$ is the time delay between m-the microphone in the microphone array **102** and the target speaker **200** in the l-th position. The delay represents the distance between the speaker and the microphone array **102**. In some examples, the delay and sum model **38** may be advantageously utilized for vehicles **100** with a smaller interior cabin **106**, whereas the estimation mentioned above may be advantageously utilized for vehicles with a larger interior cabin **106**. It is contemplated that the ECU **12** is configured with both options, and the speech enhancement system **10** may determine the preferred option based on the configuration of the vehicle **100**. All H.sub.l(k) vectors define a direction of the target zone. The blocking matrix **40** may be built by concatenating H.sub.l(k), which represents one position within the target zone, from all of the positions within the target zone Z.sub.1-Z.sub.n. For example, the blocking matrix **40** may be built to block a direction of each column of A(k) in the following exemplary equation (c):

$$[00003] \ A(k) = [H_1(k) \ H_2(k) \ \dots \ H_L(k)] \quad (c)$$

where L is the total number of positions that were recorded within the zones Z.sub.1-Z.sub.n. If the number of positions, L, is larger than P, the L columns of matrix A(k) may be compressed to P columns, for the following null space extraction.

[0053] The blocking matrix **40** is designed to block the direction of the target signal **104a** and can be constructed in a null space **48a**. The null space **48a** may be extracted by a singular value decomposition (SVD) **48** of columns of A(k)A(k).sup.H using the following exemplary equation (d):

$$[00004] \ A(k)A(k)^H = U \ \text{Math.} \ V^T \quad (d)$$

where the blocking matrix **40** is represented by the null space of the space spanned by the columns A(k)A(k).sup.H, which represent the direction of the target signal **104a**. The directional activity application may, thus, utilize SVD **48** for a signal null value **48a** to obtain the composition of the columns. SVD **48** may be utilized by the directional activity application **14** as one example method of determining the directivity of the target signal(s) **104a**. The columns of the blocking matrix **40** are the last N=M-P columns of V, which provide an orthonormal basis of null(A(k)A(k)H.sup.H).

[0054] The energy detector **24** may then be applied to calculate the energy level for both the reference microphone **102a** and the blocked target signal **104a**, as generally described above with respect to FIG. **6**. The energy detector **24** tracks the noise floor **28** to identify a high delta change. For example, the energy levels may be calculated using the following exemplary equations (e), (f):

$$[00005] \ \Delta_{in} = S_{\text{fast}}\{Y_{in}\} - S_{\text{slow}}\{Y_{in}\} \quad (e)$$

$$\Delta_{BM} = S_{\text{fast}}\{\max(Y_{BM})\} - S_{\text{slow}}\{\max(Y_{BM})\} \quad (f)$$

where S is a smoothing operator.

[0055] The directional mask **22** is applied according to the following exemplary activity conditions (a)-(c):

$$[00006] \ \Delta_{in} > \text{EngTh} \quad (a) \quad \Delta_{BM} < \text{EngTh} \quad (b) \quad \frac{\Delta_{in}}{\Delta_{BM}} > \text{EngRatioTh} \quad (c)$$

The optimal energy threshold **26** is generated using a Monte Carlo simulation **46**, such that the energy threshold **26** is defined per audio task. For example, the identified optimal energy threshold **26** may be tailored for an audio task. While described with respect to the Monte Carlo simulation

46, it is contemplated that other methods of identifying the energy threshold 26 may be utilized. For example, in some instances the speech enhancement system 10 may be configured to focus on refining precision and may, thus, maximize the ratio. In other instances the speech enhancement system 10 may be configured to focus on noise reduction and may, thus, minimize the false negative detections. Thus, the directional activity application 14, of the speech enhancement system 10, may execute a number of simulations for various energy thresholds 26 and determine which of the simulated energy thresholds 26 are optimal.

[0056] For example, an estimation of the steering vector 52 of the target signal 104a may be based on true/false bins that are predicted positive. The speech enhancement system 10 may identify active bins of the directional mask 22 and may update the RTF 32 based on the identified bins. In some examples, the requirement is to maximize the precision, defined as the following:

$$[00007] \frac{TP}{TP+FP}$$

For the task of noise reduction, the speech enhancement system 10 wants to avoid self-cancellation of the target signal 104a. Thus, the speech enhancement system may be configured to minimize the false negative scenarios. Depending on the task, the speech enhancement system 10 is configured to optimize the energy threshold 26 based on the task. For example, in some instances the speech enhancement of a specific zone Z.sub.1-Z.sub.n, so the speech enhancement system 10 may only collect the audio signals 104 corresponding to that specific zone Z.sub.1-Z.sub.n. Thus, the optimization criteria may change depending on the task. An exemplary block diagram of the true/false bins is outlined in Table 1 below:

TABLE-US-00001 TABLE 1 True/False Bins Actual: Positive Actual: Negative Predicted: Positive TP FP Predicted: Negative FN TN

Where TP is a true positive, FP is a false positive, FN is a false negative, and TN is a true negative.

[0057] Although the speech enhancement system 10 is described herein with respect to focusing on a single zone Z.sub.1-Z.sub.n, among the plurality of zones Z.sub.1-Z.sub.n, it is contemplated that the processes described herein may be extended as a multivariate problem for multiple zones Z.sub.1-Z.sub.n. Thus, the directional activity application 14 may be utilized and scaled for multiple zones Z.sub.1-Z.sub.n within the vehicle 100, such that the entire speech enhancement system 10 is scalable. For example, the speech enhancement system 10 may apply the application 16 for each zone and compare the energy ratio of each of the zones Z.sub.1-Z.sub.n with one another to identify the dominant zone Z.sub.1-Z.sub.n.

[0058] Referring to FIGS. 7 and 8, an exemplary flow diagram for the speech enhancement system 10 is depicted. At 500, the speech enhancement system 10 records clean audio signals 104. The speech enhancement system 10 may then build, at 502, the blocking matrix 40. The speech enhancement system 10 may then transform, at 504, the audio signals 104 and may project, at 506, the blocking matrix 40 onto the audio signals 104. The speech enhancement system 10 may then determine, at 508, an absolute value of each blocked audio signal 104 per time frequency bin, and may select, at 510, a maximum value per time frequency bin.

[0059] The speech enhancement system 10 may then detect, at 512, energy changes and may determine, at 514, a ratio between energy detectors 24. The speech enhancement system 10 generates, at 516, a directional mask 22. The speech enhancement system 10 may then update, at 518, the RTF 32.

[0060] A number of implementations have been described. Nevertheless, it will be understood that various modifications may be made without departing from the spirit and scope of the disclosure. Accordingly, other implementations are within the scope of the following claims.

[0061] The foregoing description has been provided for purposes of illustration and description. It is not intended to be exhaustive or to limit the disclosure. Individual elements or features of a particular configuration are generally not limited to that particular configuration, but, where applicable, are interchangeable and can be used in a selected configuration, even if not specifically shown or described. The same may also be varied in many ways. Such variations are not to be

regarded as a departure from the disclosure, and all such modifications are intended to be included within the scope of the disclosure.

Claims

1. A computer-implemented method when executed by data processing hardware causes the data processing hardware to perform operations comprising: generating a blocking matrix based on pre-recorded signals from a target zone; receiving, at a voice activity detector, audio frames from a microphone array; applying the blocking matrix to one or more zones within a vehicle; detecting signals from unblocked zones of the vehicle; determining an activity of a target signal based on the detected signals from the unblocked zones; estimating, by a beamformer, a relative transfer function (RTF) vector based on the received audio frames and the determined activity of the target signal.
2. The method of claim 1, wherein the operations further include defining a blocking area of the blocking matrix.
3. The method of claim 1, wherein the operations further include tracking a noise floor with an energy detector.
4. The method of claim 1, wherein the operations further include generating an energy threshold by Monte Carlo simulation.
5. The method of claim 4, wherein the operations further include identifying an optimal energy threshold based on the energy threshold generated by the Monte Carlo simulation.
6. The method of claim 5, wherein the operations further include tailoring the identified optimal energy threshold for an audio task.
7. The method of claim 1, wherein generating the blocking matrix includes recording clean signals from each zone.
8. The method of claim 1, wherein the operations further include enhancing the received audio frames by transforming the audio frames into a short-time Fourier transform (STFT) domain.
9. A speech enhancement system comprising: data processing hardware; and memory hardware in communication with the data processing hardware, the memory hardware storing instructions that when executed on the data processing hardware cause the data processing hardware to perform operations comprising: generating a blocking matrix based on pre-recorded signals from a target zone; receiving, at a voice activity detector, audio frames from a microphone array; applying the blocking matrix to one or more zones within a vehicle; detecting signals from unblocked zones of the vehicle; determining an activity of a target signal based on the detected signals from the unblocked zones; and estimating, by a beamformer, a relative transfer function (RTF) vector based on the received audio frames and the determined activity of the target signal.
10. The speech enhancement system of claim 9, wherein the operations further include defining a blocking area of the blocking matrix.
11. The speech enhancement system of claim 9, wherein the operations further include tracking a noise floor with an energy detector.
12. The speech enhancement system of claim 9, wherein the operations further include generating an energy threshold by Monte Carlo simulation.
13. The speech enhancement system of claim 12, wherein the operations further include identifying an optimal energy threshold based on the energy threshold generated by the Monte Carlo simulation.
14. The speech enhancement system of claim 13, wherein the operations further include tailoring the identified optimal energy threshold for an audio task.
15. The speech enhancement system of claim 9, wherein generating the blocking matrix includes recording clean signals from each zone.
16. The speech enhancement system of claim 9, wherein the operations further include enhancing

the received audio frames by transforming the audio frames into a short-time Fourier transform (STFT) domain.

17. A speech enhancement system for a vehicle, the speech enhancement system comprising: data processing hardware; and memory hardware in communication with the data processing hardware, the memory hardware storing instructions that when executed on the data processing hardware cause the data processing hardware to perform operations comprising: generating a blocking matrix based on a steering vector, the blocking matrix including a mask; receiving, at a voice activity detector, audio frames from a microphone array; applying the blocking matrix to one or more zones within the vehicle; detecting signals from unblocked zones of the vehicle; and determining an activity of a target signal based on the detected signals.

18. The speech enhancement system of claim 17, wherein the operations further include calculating a ratio of energy changes between a reference microphone signal and a maximum value of outputs of the blocking matrix.

19. The speech enhancement system of claim 18, wherein the operations further include generating the mask of the blocking matrix based on the ratio of energy changes.

20. The speech enhancement system of claim 19, wherein the operations further include identifying active bins of the mask and updating a relative transfer function (RTF) based on the identified active bins.
