



- (51) **International Patent Classification:**
G10L 21/0208 (2013.01) *H04R 3/00* (2006.01)
- (21) **International Application Number:**
PCT/EP2022/052641
- (22) **International Filing Date:**
03 February 2022 (03.02.2022)
- (25) **Filing Language:** English
- (26) **Publication Language:** English
- (30) **Priority Data:**
2101561.5 04 February 2021 (04.02.2021) GB
2106897.8 14 May 2021 (14.05.2021) GB
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- (81) **Designated States** (*unless otherwise indicated, for every kind of national protection available*): AE, AG, AL, AM, AO, AT, AU, AZ, BA, BB, BG, BH, BN, BR, BW, BY, BZ,

CA, CH, CL, CN, CO, CR, CU, CZ, DE, DJ, DK, DM, DO, DZ, EC, EE, EG, ES, FI, GB, GD, GE, GH, GM, GT, HN, HR, HU, ID, IL, IN, IR, IS, IT, JO, JP, KE, KG, KH, KN, KP, KR, KW, KZ, LA, LC, LK, LR, LS, LU, LY, MA, MD, ME, MG, MK, MN, MW, MX, MY, MZ, NA, NG, NI, NO, NZ, OM, PA, PE, PG, PH, PL, PT, QA, RO, RS, RU, RW, SA, SC, SD, SE, SG, SK, SL, ST, SV, SY, TH, TJ, TM, TN, TR, TT, TZ, UA, UG, US, UZ, VC, VN, WS, ZA, ZM, ZW.

- (84) **Designated States** (*unless otherwise indicated, for every kind of regional protection available*): ARIPO (BW, GH, GM, KE, LR, LS, MW, MZ, NA, RW, SD, SL, ST, SZ, TZ, UG, ZM, ZW), Eurasian (AM, AZ, BY, KG, KZ, RU, TJ, TM), European (AL, AT, BE, BG, CH, CY, CZ, DE, DK, EE, ES, FI, FR, GB, GR, HR, HU, IE, IS, IT, LT, LU, LV, MC, MK, MT, NL, NO, PL, PT, RO, RS, SE, SI, SK, SM, TR), OAPI (BF, BJ, CF, CG, CI, CM, GA, GN, GQ, GW, KM, ML, MR, NE, SN, TD, TG).

Published:

— with international search report (Art. 21(3))

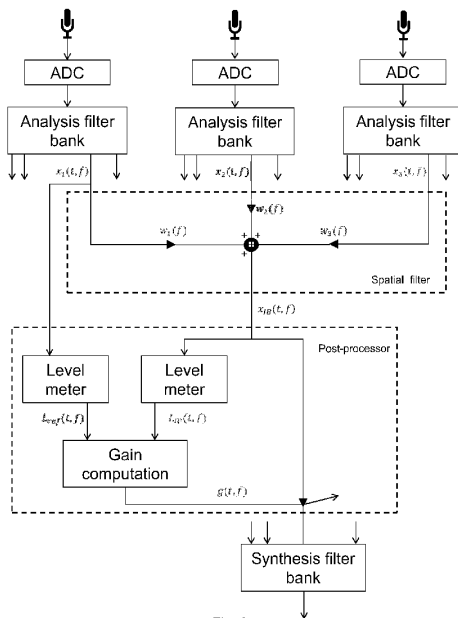
(54) **Title:** AUDIO PROCESSING

Fig. 2

(57) **Abstract:** A computer-implemented method of processing an audio signal. The method comprises: receiving from two or more microphones, respective audio signals; deriving a plurality of time-frequency signals from the received audio signals, indexed by frequency, and for each of the time-frequency signals: determining in-beam components of the audio signals; and performing post-processing of the received audio signals, the post-processing comprising: computing a reference level based on the audio signals; computing an in-beam level based on the determined in-beam components of the audio-signals; computing a post-processing gain to be applied to the in-beam components from the reference level and in-beam level; and applying the post-processing gain to the in-beam components.