WAcouSense

MCU FW PDM MIC related commands

```
mic <dB> [<Fs> [<pattern>]]
       This command sets the amplification/attenuation <db> on PDM2PCM filter,
       the sample rate <Fs> and live MIC vs. play pattern from buffer <pattern>
       db>
             0 : off (MIC disabled)
              1..52: amplification in dB: as dB - 1: 1 is 0dB
              1xx : xx = 0..20: attenuation as -xx dB
       <Fs>
             0: (default) 48 Khz
             1:32 KHz
             2:24 KHz
             3:16 KHz (results in mono, why?)
             4: 8 KHz
       <pattern>
             0: (default) live PDM MIC signal
              1 : PCM sine wave (after/without PDM filter)
             2 : PDM rectangle signal
             3 : PDM sine signal
             4 : PDM replay prerecorded PDM sine
             5 : replay captured (recorded) live signal (see mics)
       Examples:
             mic 30
                                  #live PDM signal, 30dB amplification, 48KHz
                                  #live PDM signal, 30dB amplification, 32KHz
             mic 30 1
             mic 1 0 3
                                  #play PDM sine wave (must be 48KHz only)
micc [<HP> [<filter>]]
       This command sets filter parameters for PDM2PCM filter < HP> and selects which filter to use
       <filter>
       <HP> (only for PDM2PCM filter)
             0: (default) HP filter off
              1 : coefficient 1.0 (like off)
             2 : coefficient 0.8 (strong HP filter, cuts of low frequencies below 100Hz)
             3 : coefficient 0.9
             4: coefficient 0.98
             5 : coefficient 0.995 (almost like off)
```

<filter>

- 0: own PDM filter
- 1: PDM2PCM filter, without Post Filter
- 2 : own PDM filter (the same as 0)
- 3 : PDM2PCM filter, with Post Filter (additional LP filter to remove above 10KHz)

micv <v1> <v2>

This command sets filter volume scaling factors: **<v1>** is used to convert int to float before filters are applied (input scaling), **<v2>** scales up the int values after filtering (output scaling). The v1 and v2 have just an effect if: a) the PDM2PCM **Post Filter** is enabled or b) "own PDM Filter" is used.

Typical examples for **Post Filter**:

```
micv 1 27520 #default
micv 2 13750
micv 4 6850
micv 8 3420
micv 16 1710
```

In case "own PDM Filter" is used:

v1 is integer multiplier to upscale results (output only), v2 is factional part, taken as v2 / 100000.

If both are zero (omitted) – signal is completely muted.

mics

This command captures a period of a real PDM MIC live signal.

It works only for 48KHz Fs.

This capture (snapshot) can be replayed with command:

```
mic <db> 0 5 #play (looping) recorded capture
```

Remark:

This command prints a lot of hex values on UART. This is intended to "copy and paste" as samples into C source code (see pattern as 4), as a replay pattern.

Replaying recorded capture or pattern with a different **Fs>** will result in a pitch change.

Syntax explanation:

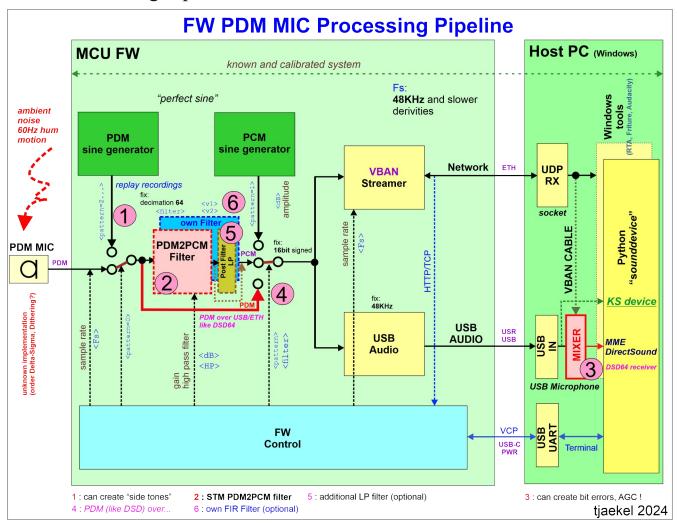
```
<dB>: a value, as mandatory, as an unsigned long (just positive) value, e.g. 3
```

[\dots] : optional parameters, when omitted taken as 0 (default)

Parameters can be omitted if they are on the most right hand side. A parameter as 0 on the left hand side with a following parameter not zero - it must be provided as 0.

The signs <> [] used here are not part of a command. It just illustrates what is mandatory or optional.

Audio Processing Pipeline and effects of command parameters



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