

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
mic 30 1        #live PDM signal, 30dB amplification, 32KHz
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 fractional 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

[...] : 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.

FW PDM MIC Processing Pipeline

MCU FW (known and calibrated system)

Host PC (Windows)

ambiant noise
60Hz hum
motion

PDM MIC

PDM sine generator

PCM sine generator

replay recordings
fix: decimation 64
<v1>
<v2>

own Filter

PDM2PCM Filter

Post Filter LP

fix: 16bit signed

amplitude

sample rate
<Fs>

fix: 48KHz

VBAN Streamer

Network (ETH)

UDP RX (socket)

Windows tools (RTA, Friture, Audacity)

Python "sounddevice"

KS device

Mixer

USB Audio

USB IN

USB Microphone

USB UART (Terminal)

FW Control

sample rate
<Fs>

gain high pass filter
<dB>
<HP>

PDM over USB/ETH like DSD64

fix: 48KHz

fix: 16bit signed

sample rate
<Fs>

fix: 48KHz

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