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# Data Article

# Dataset for Bluetooth 5.1 Direction of Arrival with non Uniform Rectangular Arrays



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#### a r t i c l e i n f o

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### a b s t r a c t

This paper presents a dataset for Bluetooth 5.1 direction of arrival (DoA). The dataset was generated with a specifically designed mathematical model of a non-uniform rectangular antenna array. The Python source files that generated the dataset are also provided. The dataset was conceived as a starting point for developing and validating DoA algorithms for real-life scenarios. Unlike other datasets, it contains Bluetooth signals with not only varying intensity of additive white Gaussian noise, but also coherent interfering signals with random DoA coordinates. The dataset is divided into two branches, one consisting of pure sinusoidal tones and the second comprised of baseband Bluetooth signals. Since the codebase which generates the data is included, this dataset has a high reuse potential, and it can be modified to suit also other types of signals or different array topologies. © 2021 Published by Elsevier Inc.

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#### <span id="page-1-0"></span>**Specifications Table**



# **Value of the Data**

This dataset simulates the interaction of Bluetooth 5.1 signals with the URA8 array topology. The latest Bluetooth 5.1 standard has introduced important features to aid direction-finding solutions [\[1\].](#page-16-0) URA8 is an antenna array composed of eight elements equally spaced on a square edge. This type of array is thoroughly analyzed in  $[2]$  and is shown in Fig. 1. The mathematical model of the array is presented in detail in [Section](#page-7-0) 3.1.

• The dataset described in this paper is useful because it provides a valuable tool for the development and validation of DoA algorithms for Bluetooth 5.1 signals. It contains coherent interference with respect to the useful signal, which is always present in real-life scenarios. Since DoA algorithms are usually tested and compared on sinusoidal signals, both sinusoidal signals and Bluetooth signals are included in this dataset. Moreover, the dataset provides the signals in the form of IQ samples, which is how they are outputted by modern Bluetooth 5.1 devices.



**Fig. 1.** Geometrical configuration of the non-uniform rectangular array with 8 patch elements.

- <span id="page-2-0"></span>• Anyone who develops Bluetooth 5.1 tracking applications can benefit greatly from this dataset, because the simulated data are comparable to those obtained through actual measurements. The data have been compared with real-world IQ samples generated by an STMicroelectronics transceiver prototype, fully compliant with the Bluetooth 5.1 standard. This prototype has been equipped with an eight-sensors patch antenna array, with the same topology as the mathematical model of the array shown in [Fig.](#page-1-0) 1 and described in [Section](#page-7-0) 3.1. Experimental results have shown that the measured IQ samples match the signals modeled in the presented dataset with a signal to noise ratio (SNR) of 60dB and a with signal to interference ratio (SIR) below 60dB, showing a mean error equal to 0 and a standard error deviation around 1%. The non-zero value of the standard error deviation is due to a residual stochastic uncompensated frequency offset on the measured IQ samples.
- By including the Python codebase that generates the dataset alongside the dataset itself, the data can be modified and expanded at wish. For example, more than two interfering signals could be added, the topology of the array could be modified, or the number of simulated measurements could be increased. The latter aspect is especially valuable when considering the training of artificial neural networks, which require large quantities of data to generalize appropriately.

#### **1. Data Description**

The dataset consists of .csv data, along with .py Python source files that are used to generate the data. It is divided into two branches, each one corresponding to the type of signal which interacts with the rectangular array. The first branch is devoted to radio frequency (RF) sinusoidal signals, while the second one contains Bluetooth 5.1 signals. Each branch is selfcontained within its own folder. The content of the PureTone\_data/ and BLE\_5\_1\_data/ folders is described in Sections 2.1 and [2.2](#page-6-0) respectively. The Python code that was used to generated the dataset, contained in folders PureTone\_code/ and BLE\_5\_1\_code/, is described in [Section](#page-9-0) 3.3.



#### *1.1. RF 2.4GHz pure tone branch*

The directory structure is shown below. The data is organized in a series of simulated tests, each contained within its own folder, whose name always begins with Test\_\*/. Each test folder then contains  $N_{\phi} \cdot N_{\theta}$  sub-folders, where  $N_{\phi}$  and  $N_{\theta}$  are the number of possible azimuth and elevation angles of the useful signal, respectively. The description of every dataset parameter is reported in [Table](#page-3-0) 1, and is valid for both dataset branches. The range of each PureTone dataset branch parameter is reported in [Table](#page-3-0) 2.

# <span id="page-3-0"></span>**Table 1**





**Table 2** Dataset parameters value range for the PureTone branch.

| Parameter   | Value range   | Measure unit |
|---|---|--------------|
| $$  | ${0, 2}$  | MHz          |
| $0$ , $$  | ${6, 12}$   | dB           |
| $\langle d_i \rangle$                                 | ${12, 30, 60}$  | dB           |
| $\langle \phi_i \rangle$ , $\langle \theta_i \rangle$ | $\phi_i \in [0, 359]$ (1° step), $\theta_i \in [0, 90]$ (3° step)                       | $\circ$      |
| $<\!\!\phi_i^1\!\!>,\ <\!\!\theta_i^1\!\!>$           | Same as $\langle \phi_i^0 \rangle$ , $\langle \theta_i^0 \rangle$ , random distribution | $\circ$      |
| $<\!\phi^2\!\!>,\; <\!\theta^2\!\!>$                  | Same as $\langle \phi_i^0 \rangle$ , $\langle \theta_i^0 \rangle$ , random distribution | $\circ$      |

```
PureTone_data/
    \_Test\_Iof<a_1>MHz\_SIRs<b_1>dB<c_1>dB\_SNR<d_1>dB
    \verb|Test_Iof<a_2>MHz_SIRs<a_2>dB<a_2>dB\_SNR<a_2>dB\verb|Test_Iof<a_3>MHz_SIRs&lt;b_3>dB&lt;c_3>dB_SNR&lt;d_3>dB\text{\texttt{Test\_Iof}} \leq a_N>MHz_SIRs\text{<} b_N>dB\text{<} c_N>dB_SNR\text{<} d_N>dB
         _sig_azim0_elv0
         _sig_azim0_elv3
         _sig_azim0_elv6
          sig_azim0_elv87
         _sig_azim1_elv0
         _sig_azim1_elv3
         _sig_azim1_elv6
           \cdots_sig_azim1_elv87
         _sig_azim2_elv0
         sig_azim2_elv3_
         _sig_azim2_elv6
          \text{sig_axim} \leq \phi_i \geq \text{elv} \leq \theta_isig_azim359_elv87
               _doares_saz359_sel87.csv
                info_saz359_selv87_<hh>_<mm>_of_<DD>_<MM>_<YYYY>.csv
               \angle X_m<n<sub>0</sub>>_saz359_sel87_iaz[<\phi_1^1> <\phi_1^1>]_iel[<\theta_1^1> <\theta_1^2).csv
              \underline{\hspace{0.3cm}} X_m<n<sub>0</sub>>_saz359_se187_iaz[<\phi_2^1> <\phi_3^1>]_iel[<\theta_2^1> <\theta_3^2>].csv
               \pm X_m<n<sub>0</sub>>_saz359_sel87_iaz[<\phi_3^1> <\phi_4^1>]_iel[<\theta_3^1> <\theta_3^2>].csv
              \pm X_m<n<sub>1</sub>>_saz359_se187_iaz[<\phi_4^1> <\phi_4^1>]_iel[<\theta_4^1> <\theta_4^2)].csv
                \mathbb{E} \mathbb{X}_1 \mathbb{M} \leq n_1 > \mathbb{E} \mathbb{R}az359_se187_iaz[<\phi^1_s> <\phi^1_s>]_iel[<\theta^1_s> <\theta^2_s>].csv
               \mathbb{Z} \times \mathbb{R} \leq n_1 > \mathbb{Z}saz359_sel87_iaz[<\phi^1_{6} > \phi^1_{6} > \mathbb{Z}iel[<\theta^1_{6} > \phi^2_{6} > \mathbb{Z}.csv
```
There are three types of .csv files for each  $sig\_azim < \phi_i > _{e}lv < \theta_i$  folder, all of them using UTF-8 encoding and comma as separator. **The following description is valid for both dataset branches:**

- doares\_saz $\langle \phi_i \rangle$  sel $\langle \theta_i \rangle$ . csv files contain the DoA of the useful signal, estimated by the Multiple signal classification algorithm (MUSIC), which is described in [Section](#page-9-0) 3.2. They are meant to serve as a ground truth for performance comparison with other DoA algorithms. An example of such file is shown in [Table](#page-5-0) 3. The column headers and their abbreviations correspond to the parameters described in [Table](#page-3-0) 1. The results obtained by MUSIC are contained in res\_azim (resulting azimuth) and res\_elev (resulting elevation) columns.
- info\_saz $\langle \phi_i \rangle$  = sel $\langle \theta_i \rangle$  =  $\langle$ hh $\rangle$ \_ $\langle$ mm $\rangle$ \_of\_ $\langle$ DD $\rangle$ \_ $\langle$ MM $\rangle$ \_ $\langle$ YYYY $\rangle$ .csv files are useful in the case that one does not want to rely on the hierarchy of the directory tree to read the key dataset parameters. Their generic structure is shown in [Table](#page-5-0) 4. They contain the useful signal angular position range (minimum, maximum, and step), the position of the useful signal, the frequency offset  $\langle a_i \rangle$ , the signal to noise ratio  $\langle d_i \rangle$  and the first interfering signal to useful signal ratio  $\langle b_i \rangle$ . The parameters correspond to those described in [Table](#page-3-0) 1.

<span id="page-5-0"></span>**Table 3** Example of a do ${\tt ares\_saz} \! <\! \phi_i \! > \! \_ {\tt sel} \! < \! \theta_i \! > \! \ldots$ csv file.

| meas | sig_azim | sig_elev | <b>CTE</b> | iaz.      | iel     | res azim   | res elev |
|------|----------|----------|------------|-----------|---------|------------|----------|
| 0    | 0.000000 | 0.000000 | [0 0]      | [305 16]  | 16 44]  | 324.000000 | 9.000000 |
|      | 0.000000 | 0.000000 | [0 0]      | [305 16]  | 16 441  | 321.000000 | 5.000000 |
| 0    | 0.000000 | 0.000000 | 11 11      | [292 169] | [44 44] | 292.000000 | 6.000000 |
|      | 0.000000 | 0.000000 | 11 11      | [292 169] | [44 44] | 273.000000 | 9.000000 |
| 0    | 0.000000 | 0.000000 | 11 11      | [337 355] | [21 33] | 329.000000 | 9.000000 |
|      | 0.000000 | 0.000000 | '1 11      | [337 355] | [21 33] | 345.000000 | 9.000000 |

### **Table 4**

Generic content of  $info\_saz < \phi_i > _sel < \theta_i > _s< hh > _s< mm > _oof _< LDD > _s< MM > _s< YYYY > .$  csv files.



<span id="page-6-0"></span>• X\_m< *ni* >\_<sup>∗</sup>.csv files contain the <sup>8</sup> <sup>×</sup> <sup>70</sup> *XIQ* complex-valued IQ matrices, which result from the interaction of the signals and the rectangular antenna array. These files do not contain any header. Each  $8 \times 1$  column is a complex-valued IQ vector  $\mathbf{x}_{I0}$ <sup>T</sup>. The complex numbers are in the format shown in Table 5

#### **Table 5**

Example of a complex-valued IQ vector  $\mathbf{x}_{I0}^T$  contained in  $X_m < n_i >$ <sup>\*</sup>. csv files.



#### *1.2. Baseband Bluetooth 5.1 signal branch*

The directory structure is very similar to the PureTone branch and is reported below. The difference consists in the greater number of parameters due to the presence of the residual frequency offset and the constant tone extension (CTE), which are typical of Bluetooth 5.1 signals. [Table](#page-3-0) 1 provides a description for each parameter, while Table 6 reports the value range that

**Table 6** Dataset parameters value range for the BLE\_5\_1 branch.

| Parameter   | Value range   | Measure unit |
|---|---|--------------|
| $$  | {0}   | <b>MHz</b>   |
| $\langle b_i \rangle$ , $\langle c_i \rangle$             | $\{3, 6, 12, 18, 24, 30, 36, 48, 54, 60\}$  | dB           |
| $\langle d_i \rangle$                                     | $\{30, 48, 60\}$  | dB           |
| $\langle \hat{e_i} \rangle$                               | $\{-3, 0, +3\}$   | kHz          |
| $\langle e_i \rangle$                                     | $e_i \in [-3, +3]$ , Gaussian distribution  | hHz          |
| $\langle f_i \rangle$                                     | ${1200}$  | $\mu$ s      |
| $\langle g_i \rangle$ , $\langle h_i \rangle$             | ${0, 1}$ (OFF/ON)   |              |
| $-p_i$  | {1}   | Mbit/s       |
| $\langle q_i \rangle$                                     | ${1, 2}$  | $\mu$ s      |
| $\langle \phi_i \rangle$ , $\langle \theta_i \rangle$     | $\phi_i \in [0, 359]$ (1° step), $\theta_i \in [0, 90]$ (3° step)                                   | $\circ$      |
| $\langle \phi_i^1 \rangle$ , $\langle \theta_i^1 \rangle$ | Same value range as $\langle \phi_i^0 \rangle$ , $\langle \theta_i^0 \rangle$ , random distribution | $\circ$      |
| $<\!\phi_i^2\!\!>,\; <\!\theta_i^2\!\!>$                  | Same value range as $\langle \phi_i^0 \rangle$ , $\langle \theta_i^0 \rangle$ , random distribution | $\circ$      |

<span id="page-7-0"></span>can be assumed by each parameter. The three types of .csv files within each  $sig\_azim <$  $\phi_i$ >\_elv<  $\theta_i$ > folder are identical with respect to the PureTone dataset branch.<br>BLE\_5\_1\_data/

 $\Box$ DR< $p_i$ >Mbps\_Switching< $q_i$ >us  $\_ Test\_Iof >MHz $\_SIRS **b_1**$ >dB< $c_1$ >dB $\_SNR **d_1**$ >dB $\_rfo **e_1**$ >KHz $\_CTE **f_1**$ >us$  $\_ Test\_Iof <_{a_2}$ >MHz\_SIRs< $b_2$ >dB< $c_2$ >dB\_SNR< $d_2$ >dB\_rfo< $e_2$ >KHz\_CTE< $f_2$ >us  $\_Test\_I$ of< $a_3$ >MHz $\_SIRS$ < $b_3$ >dB< $c_3$ >dB $\_SNR$ < $d_3$ >dB $\_rf$ o< $e_3$ >XHz $\_CTE$ < $f_3$ >us \_Test\_Iof< $a_N$ >MHz\_SIRs< $b_N$ >dB< $c_N$ >dB\_SNR< $d_N$ >dB\_rfo< $e_N$ >XHz\_CTE< $f_N$ >us \_sig\_azim0\_elv0 sig\_azim0\_elv3 sig\_azim0\_elv6  $\cdots$ sig\_azim0\_elv87 \_sig\_azim1\_elv0 sig\_azim1\_elv3 sig\_azim1\_elv6  $\overline{\phantom{a}}$ sig\_azim1\_elv87 \_sig\_azim2\_elv0 sig\_azim2\_elv3 sig\_azim2\_elv6  $\ldots$  $_s$  sig\_azim< $\phi_i$ >\_elv< $\theta_i$ > sig\_azim359\_elv87 \_doares\_saz359\_se187.csv info\_saz359\_selv87\_<hh>\_<mm>\_of\_<DD>\_<MM>\_<YYYY>.csv \_X\_mO\_saz359\_sel87\_cte[<g<sub>l</sub>> <h<sub>1</sub>>]\_iaz[< $\phi_1^1$ > < $\phi_1^1$ )\_iel[< $\theta_1^1$ > < $\theta_1^2$ )\_roff< $m_1$ >hHz.csv  $\underline{\phantom{a}}$  X\_mO\_saz359\_sel87\_cte[<g<sub>2</sub>> <h<sub>2</sub>>]\_iaz[< $\phi_2^1$ > < $\phi_3^1$ >]\_iel[< $\theta_3^1$ > < $\theta_2^2$ >]\_roff<m<sub>2</sub>>hHz.csv \_X\_mO\_saz359\_se187\_cte[<g<sub>3</sub>> <h<sub>3</sub>>]\_iaz[< $\phi_3^1$ > < $\phi_4^1$ >]\_ie1[< $\theta_3^1$ > < $\theta_4^2$ >]\_roff< $m_3$ >hHz.csv \_ X\_m1\_saz359\_se187\_cte[< $g_4$ > < $h_4$ >]\_iaz[< $\phi^1_A$ > < $\phi^1_A$ >]\_iel[< $\theta^1_A$ > < $\theta^2_A$ >]\_roff< $m_4$ >hHz.csv \_X\_m1\_saz359\_sel87\_cte[< $g_5$ > < $h_5$ >]\_iaz[< $\phi^1_\varsigma$ > < $\phi^1_\varsigma$ >]\_iel[< $\theta^1_\varsigma$ > < $\theta^2_\varsigma$ >]\_roff< $m_5$ >hHz.csv \_X\_m1\_saz359\_sel87\_cte[<g<sub>6</sub>> <h<sub>6</sub>>]\_iaz[< $\phi_6^1$ > < $\phi_6^1$ )\_iel[< $\theta_6^1$ > < $\theta_6^2$ >]\_roff<m<sub>6</sub>>hHz.csv

#### **2. Experimental Design, Materials and Methods**

The dataset is based on a mathematical model of a non-uniform rectangular antenna array with 8 elements. The following sections provide a description of the mathematical model of the array, of the impinging signals, and of their Python implementation which is provided with the dataset.

#### *2.1. Mathematical model*

<https://www.overleaf.com/project/617a8bcbbeb847848f1fd78c> The array model corresponds a rectangular array of  $N = 8$  numbered antenna elements, evenly spaced at a distance  $d = \frac{\lambda}{2.5}$  on <span id="page-8-0"></span>a square edge, as shown in [Fig.](#page-1-0) 1. Because of the lack of a central element, this array is nonuniform.

The rectangular array receives *M* signals  $s_i(n)$ , incident with angles

$$
(\theta_1, \phi_1), (\theta_2, \phi_2), ..., (\theta_m, \phi_m), ..., (\theta_M, \phi_M)
$$
\n(1)

where  $\theta_m$  and  $\phi_m$  are the *m*-th elevation and azimuth angles of each signal. The signals have also a defined power  $P_1$ ,  $P_2$ ,  $P_m$ , ...,  $P_M$ .

The array response matrix *A* describes the interaction of the impinging signals with the rectangular antenna array. For the topology of the array shown in [Fig.](#page-1-0) 1, the array response matrix is

$$
A = [\boldsymbol{a}(\theta_1, \phi_1)^T, \boldsymbol{a}(\theta_2, \phi_2)^T, ..., \boldsymbol{a}(\theta_M, \phi_M)^T)]
$$
\n(2)

where the array response vectors  $\mathbf{a}(\theta, \phi)$  are defined as:

$$
\mathbf{a}(\theta,\phi) = \begin{bmatrix} 1 \\ e^{-j\gamma(\theta)(d\cos(\phi))} \\ e^{-j\gamma(\theta)(2d\cos(\phi))} \\ e^{-j\gamma(\theta)(2d\cos(\phi) + d\sin(\phi))} \\ e^{-j\gamma(\theta)(2d\cos(\phi) + 2d\sin(\phi))} \\ e^{-j\gamma(\theta)(d\cos(\phi) + 2d\sin(\phi))} \\ e^{-j\gamma(\theta)(d\sin(\phi))} \\ e^{-j\gamma(\theta)(d\sin(\phi))} \end{bmatrix} \tag{3}
$$

with 
$$
d = \frac{\lambda}{2.5}
$$
 and  
\n $\gamma(\theta) = 2\pi \sin(\theta)$  (4)

As mentioned in [Section](#page-2-0) 2, the dataset is subdivided into two branches. For the **Bluetooth 5.1 dataset branch**, the signals  $s_i(n)$  are a baseband model of the signals at the analog to digital converter (ADC) of the Bluetooth receiver. Apart from the frequency offset, the carrier translation and the RF impairments are negligible for the study of the DoA. Regarding the frequency offset between the transmitter and the receiver, the majority of the offset is corrected at the automatic frequency corrector (AFC). However, a small part of this offset typically remains, which we indicate as  $f_{off}$ . This residual offset is able to impact negatively the estimation of the DoA. Our mathematical model of BLE 5.1 signals is that of binary Gaussian frequency shift keying (GFSK) modulated signals:

$$
s_i(n) = e^{j \cdot \alpha(n)} \cdot e^{(2 \cdot \pi \cdot f_{off} \cdot n)/F_s}
$$
 (5)

$$
\alpha(n) = \frac{2\pi \cdot F_{dev}}{F_s} \cdot \sum_{k=0}^{n} \sum_{i} b_i p(k - iN_s)
$$
\n(6)

$$
F_{dev} = \frac{DR \cdot h}{2} \tag{7}
$$

where *DR* is the data rate,  $f_{off}$  is a possible frequency offset error,  $F_s = 16MHz$  is the sampling frequency at the receiver's ADC, *n* represents the discrete-time index, *h* is the modulation index,  $p(n)$  is the symbol pulse as defined in [\[3\],](#page-16-0) and  $b_i \in \{1, -1\}$  is the binary symbol to be transmitted. As a mathematical model for the baseband CTE, Eq. 5 with all binary symbols *bi* equal to 1 is used.

The baseband samples at the Bluetooth receiver, running at a proper sample period  $T_s$  without ADC impairments can be expressed as:

$$
X_{T_5} = A \cdot \begin{bmatrix} P_1 & 0 & \dots & 0 \\ 0 & P_2 & \dots & 0 \\ 0 & \dots & \dots & 0 \\ 0 & 0 & \dots & P_M \end{bmatrix} \cdot \begin{bmatrix} s_1 \\ s_2 \\ \dots \\ s_M \end{bmatrix} + \mathbf{n}
$$
 (8)

<span id="page-9-0"></span>where  $s_1, s_2, ..., s_M$  are the complex impinging signals, generated by using [Eq.](#page-8-0) 5 with a sampling period  $T_s$  and  $\boldsymbol{n} = [n_1, n_2, ..., n_8]^T$  is the noise vector at the analog to digital converter (ADC), with zero mean and variance  $\sigma^2$ , which is related to the SNR of the useful tag signal. The noise vector is comprised of AWGN noise, which approximates the noise of the receiver, and does not take into account the interfering signals. When one of the *si* signals is a CTE, all binary symbols are set to 1, otherwise they are selected randomly.

Considering the IQ sampling process of the Bluetooth receiver and calling *Tswitch* and  $T_{sample} = 1/F<sub>s</sub>$  the switch slot duration and the sampling slot duration respectively, both set to 1  $\mu$ *s* or 2  $\mu$ *s*, the  $X_{T_s}$  data are down-sampled with a factor of *round*( $(T_{switch} + T_{sample})/T_s$ ), thus obtaining the IQ samples in form of  $X_{IO}$ , which is a complex matrix with size  $N \times N_{\text{sagnn}}$ .

Because of the filtering chain, the power of the adjacent channels, alternate channels and all remaining channels is strongly attenuated in comparison to the reflections of useful signal. This means that, in the BLE 5.1 model, all of the interfering signals are in fact the reflections of the useful signal itself.

In the case of **pure tone dataset branch**, the signals  $s_i(n)$  are defined as

$$
s_i(n) = e^{j \cdot 2\pi \cdot (F_c + ch) \cdot (n/F_s)}
$$
\n(9)

where  $F_c = 2.4GHz$ , *ch* is the channel spacing (which is an integer multiple of 2*MHz*), and  $F_s$ is the sampling frequency, set to  $F_s = F_c \cdot 32$ . In this model, we do not decimate the  $s_i$  signals.

The IQ samples are obtained as

$$
X_{IQ} = A \cdot \begin{bmatrix} P_1 & 0 & \dots & 0 \\ 0 & P_2 & \dots & 0 \\ 0 & \dots & \dots & 0 \\ 0 & 0 & \dots & P_M \end{bmatrix} \cdot \begin{bmatrix} s_1 \\ s_2 \\ \dots \\ s_M \end{bmatrix} + \mathbf{n}
$$
(10)

where *n* is the AWGN noise at the antenna elements. Please note that the noise *n* which is added in the pure signal model is much more wideband when compared to the *n* noise of the Bluetooth model, because the BLE model is a baseband model, while the pure tone signal is an RF model. Therefore, the signal to noise ratios (SNRs) obtained with the two models are not comparable. This is because the SNR depends on the bandwidth of the noise channel.

#### *2.2. The MUSIC algorithm*

In addition to the modeled RF 2.4GHz and Bluetooth 5.1 signals, this dataset provides a ground truth for the development and testing of different algorithms. For this purpose, the Multiple signal classification (MUSIC) algorithm [\[4\]](#page-16-0) was implemented and applied to the simulated dataset signals. MUSIC is a signal subspace method for DoA estimation, which exploits the eigen-structure of the autocorrelation matrix of the signal to find the signal sub-space and noise sub-space. The MUSIC pseudo-spectrum  $P(\phi, \theta)$  is obtained in following way:

$$
P(\phi,\theta) = \frac{1}{\mathbf{a}(\phi,\theta)^H \cdot Q_n \cdot Q_n^H \cdot \mathbf{a}(\phi,\theta)}
$$
(11)

where  $\mathbf{a}(\theta, \phi)$  are the array response vectors defined in [Eq.](#page-8-0) 3 and  $Q_n$  the eigen vectors of the noise sub-space. The DoA estimate of the source signal is then obtained by finding the peak in the above defined pseudo-spectrum.

#### *2.3. Python implementation*

*Note on licensing.* The Python implementation that is shipped with this dataset is free software: you can redistribute it and/or modify it under the terms of the GNU General Public License as published by the Free Software Foundation, either version 3 of the License, or (at your option) any later version. If you make improvements to the code, you are invited to share those changes with the community.

*File structure and dependencies.* The Python implementation is written in Python 3 and is contained inside the following <sup>∗</sup>.py files:



 $\overline{2}$ 

 $\overline{4}$ 

5

 $\ddot{6}$ 

 $\overline{7}$ 

 $\,$  8  $\,$ 

 $\overline{Q}$ 

The following dependencies are needed: os, time, numpy, scipy and matplotlib.

The following paragraphs describe the content of each Python module by making also reference to the mathematical equations of the previous sections.

*The algorithmic module.* The algorithmics\_URA8.py module is common to both branches of the dataset. It contains the definition of the conventional steering vector for URA8 as defined in [Eq.](#page-8-0) 3 and the implementation of the MUSIC algorithm as described in [Section](#page-9-0) 3.2. The function which generates the steering vector is the following:

```
# Conventional steering vector for URA8 ref. eq. (3)
\mathbf{1}def CSV_URA8(Nant, dx, dy, apx, apy, phi, theta):
\overline{2}\overline{\mathcal{L}}gamma = np \cdot sin(theta)\overline{4}SS = np \tvert zeros((Nant)), dtype = complex)5
         for nx in range (Nant):
6
              SS[nx] = np.exp(-1j*gamma*2*np.pi*(dx*(apx[nx])*np.cos(phi)+dy*(apy[nx])*np.sin(phi))) ;
\overline{7}return SS
\bar{8}
```
The URA8 steering vector from [Eq.](#page-8-0) 3 is also defined inside the MUSIC implementation:

```
#STeering vector for URA8 ref. eq(3)
SV = np.array([1.0 + 1 j * 0.0,np. exp(-1j*gamma\_vec[idy]*(dxcos_phi[idx])),
                    np \cdot exp(-1j*gamma\_vec[idy]*(2*dxcos_phi[idx])),
                    np.\exp(-1j*\texttt{gamma\_vec}[\texttt{idy}]*(2*\texttt{dx}\texttt{cos\_phi}[\texttt{idx}]+\texttt{dysin\_phi}[\texttt{idx}])),
                    np. exp(-1j*gamma2vec[idy]*(2*dxcosphi[idx]+2*dysinphi[idx]))np. exp(-1)*gamma\_vec[idy]*(dxcos_{phi}[idx]+2*dysin_{phi}[idx]),
                    np.exp(-lj*gamma_vec[idy]*(2*dysin_phi[idx])),
                    np.\exp(-1j*gamma\_vec[idy]*(dysin_phi[idx])) ));
```
The MUSIC pseudo-spectrum from [Eq.](#page-9-0) 11 is then estimated:

```
\mathbf{1}if np.isscalar(Qn2) == True:
\gamma\overline{3}SVH = SV.com ( ) . Tnumerator = np.matmul(SVH, SV)\overline{4}Qn2SV = Qn2*SV\overline{5}denominator = np.matmul(SVH, Qn2SV)\ddot{\phantom{0}}\overline{7}\bar{8}else:
                           \texttt{SVH = SV.config().T}\alphanumerator = np.matmul(SVH, SV)10<sup>1</sup>denominator = np.matmul(SVH, np.matmul(Qn2, SV))1112
                      PP = numerator / denominator # ref. eq. (11)13
```
*The signal model modules.* For the Bluetooth 5.1 branch, the signals are modeled inside the e2e\_URA8\_BLE.py module. The frequency deviation from [Eq.](#page-8-0) 7 is implemented as

 $fdev = data_rate * mod_index / 2.0; #frequency deviation ref. eq. (7)$ 

The baseband signals  $s_i(n)$  with the frequency offset  $f_{off}$ , as defined in [Eqs.](#page-8-0) 5 and [6,](#page-8-0) are implemented as

```
freqoff = Ch_num# CHannel offset (in Hz)
                                 #numer of channels BLE
\overline{2}BB_ModulatedSignal = np. exp(1j * (2.0 * np.pi * np.cumsum(fdev * pckt_upsampled)) / (Fbb));#signal in baseband ref. eq. (5) and eq. (6)
        len_BB = np.size(BB_ModulatedSignal, 0)3
        \tt{times} = np.arange(0, len_BB)/Fbb#\overline{4}time axis
        y_{\texttt{-tmp}} = np \cdot exp(1j * 2.0 * np \cdot pi * time x * freqoff)#5
         add frequency offset. ref. eq. (5)
        BB_ModulatedSignal = y_tmp*B_mlodulatedSignal6
                                                                                                           #signal in baseband for every channel
```
The directions of the incident signals from [Eq.](#page-8-0) 1 and the array response vectors from Eq. 3 are generated in

```
# build the A matrix
                                                                            ref. eq. (1)for h in range (len(a\_phi)):
\overline{2}theta = a_{\text{t}} theta[h]*np.pi/180.0
3
\overline{4}phi = a_{phi}[h]*np.pi/180.0;\overline{5}a_{\text{amp}} = algo.CSV_URA8(N,d,d,apx,apy,phi,theta) #ref.eq. (3)
```
The array response matrix from [Eq.](#page-8-0) 8 is implemented in the following lines:

```
for meas in range (measures):
\mathbf{1}\overline{2}noise = np.sqrt(noise_energy/2)*(np.random.randn(N, samples) +1j*np.random.randn(N,
         samples)); #Uncorrelated noise
           AdiagVrx = np.matmul(A, diagVrx)\DeltaAdiagVrxsig = np.matmul(AdiagVrx, rx\_sig)X = (AdiagVrxsig+noise); # X matrix ref. eq. (8)
5
6
           X = X[:,::down\_iq]#donw-sampling the ADC signal.
                                                           #Decimation after adc
            X = X[:, \text{ gauss\_del}:] #remove the gauss filter delay
                                                      #Remove signals delayed due to Gaussian
        filter
```
For the PureTone branch, the signals are modeled inside the e2e\_URA8\_PureTone.py module. The signals  $s_i(n)$  from [Eq.](#page-9-0) 9 are implemented as follows:

```
\mathbf{I}# build the signal matrix
\gammafor idx in range(len(a_phi)):
\overline{3}# stack in a tx signal matrix all trasmitted signals
\bar{A}if (\text{idx} == 0):
\overline{5}tx\_sig = np.exp(1j*2*np.pi*Fsin[idx]*t);#test signal ref. eq. (9)
                  rx\_sig = tx\_sig\sqrt{6}\overline{\tau}8
              else:
                  tx\_sig = np.exp(1j*2*np.pi*Fsin[idx]*t);#test signal ref. eq. (9)
Q10rx\_sig = np.vstack((rx\_sig, tx\_sig))
```
The array response vector from [Eq.](#page-8-0) 3, which comprise the array response matrix from [Eq.](#page-8-0) 2 are then computed in the following lines:

```
# build the A matrix
\mathbf{I}for h in range(len(a_phi)):
\bar{2}theta = a_{\text{t}} theta[h]*np.pi/180.0
\overline{3}\overline{4}phi = a_{phi}[h]*np.pi/180.0;a_tmp = algo.CSV_URAS(N,d,d,apx,apy,phi,theta) #ref. eq.(3)
\leq\overline{6}\gammaif(h == 0):
\bar{8}A = a_{\text{imp}}\alphaelse:
10A = np \cdot vstack((A, a_tmp))
```
The  $X_{10}$  sample matrix from [Eq.](#page-9-0) 10 is then obtained:

```
A = np.transpose(A)\overline{1}Pw = 10.0** (Pw_dB/20.0)\overline{\mathcal{L}}diagVrx = np.diag(Pw)\mathcal{R}\overline{A}for meas in range(measures):
             noise = np.sqrt(noise_energy/2)*(np.random.randn(N, samples) +1+*np.random.randn(N,
\overline{5}samples)); #Uncorrelated noise
             AdiagVrx = np.matmul(A, diagVrx)6
\overline{7}AdiagVrxsig = np.matmul(AdiagVrx, rx_sig)ref. eq(10)\bar{\mathbf{x}}X = (AdiagVrxsig+noise); # X matrix
```
*The dataset generator modules.* The dataset is generated by the e2e\_URA8\_BLE\_DSgen.py and the e2e\_URA8\_PureTone\_DSgen.py modules. Both modules do not accept console arguments. The Bluetooth 5.1 dataset generator allows for the setting of the following parameters before being launched:

```
interf_Ntests = 3 # number of test with different interfers positions for each tag azimuth
\overline{1}and elevation couple.
    measures = 2 #number of measures of IQ samples tfor each Interf. tests
\alphasave\_opt = 3 #saving options
\mathcal{R}\varDeltaN = 8 # is the antenna number
\overline{5}# BLE 5.1 settings
\overline{6}\overline{7}data_rate = le6 # data rate (or the symbol rate) in bps\bar{8}t_slot_duration = 2*1e-6 # slot duration of the IQ sampling (in sec)
    CTETime = 150*N # CTE time (ref. to all antennas) in microsec
\overline{Q}10# URA settings
11SNR = 60 # SNR of the AWGN added (refered to ADC) in dB
12max_freq_offset_res= 0 #this is the resiual frequency offset between the tx tag and rx (
13 -considering that a large amount is corrected by AFC inside the BLE demodulator)
14\texttt{saz\_start} = 0 # azimuth start angle of the tag signal
15\texttt{saz-end} = 360 #azimuth end angle of tag signal
16
    saz_step= 1 #azimuth step angle of tag signal
1718
    sel\_start = 0 #elevation start angle of tag signal
1920<sup>°</sup>sel_end = 90 #elevation end angle of tag signal
    sel\_step = 3 #elevation step angle of tag signal
21\overline{22}23\,SIR1_{\texttt{tests}} = np.array([3, 6, 12, 18, 30, 48, 60, 3, 6, 12, 18, 30, 48, 54,]) #SIR is the
         Signal to Interference ration (in dBc) SIR1 is refered to the first interf.
    SIR2_tests = np.array([3, 6, 12, 18, 30, 48, 60, 6, 12, 18, 24, 36, 54, 60,]) #SIR2 is
24
         refered to the 2th interf
```
A similar approach is used for the parameters of the PureTone dataset generator:

```
N\tan p = 70 #samples of the tone
\mathbf{I}interf_Ntests = 3 # number of test with different interfers positions for each tag azimuth
\gammaand elevation couple.
    measures = 2 #number of measures of IQ samples tfor each Interf. tests
\mathcal{R}\overline{4}save\_opt = 3 #saving options
\bar{5}SNR = 60#SNR of the AWGN (dB)
\sqrt{6}\tau\texttt{saz\_start} = 0 # azimuth start angle of the tag signal
    saz_end = 360 #azimuth end angle of tag signal
\mathcal{R}\overline{9}saz_step= 1 #azimuth step angle of tag signal
10-11sel\_start = 0 #elevation start angle of tag signal
12 \overline{ }sel_end = 90 #elevation end angle of tag signal
13
    self\_step = 3 #elevation step angle of tag signal
14SIR1_tests = np.array([3, 6, 12, 18, 30, 48, 60, 3, 6, 12, 18, 30, 48, 54,]) #SIR is the
15Signal to Interference ration (in dBc) SIR1 is refered to the first interf.
    SIR2 tests = np.array([3, 6, 12, 18, 30, 48, 60, 6, 12, 18, 24, 36, 54, 60,]) #SIR2 is
16refered to the 2th interf
```
*The testing modules.* The e2e\_URA8\_BLE\_DOAview.py and e2e\_URA8\_PureTone\_ DOAview.py modules are meant as an aid for the setting of the dataset parameters prior to its actual generation. Their output is a 3D plot of the MUSIC pseudo-spectrum. The following

code is taken from the e2e\_URA8\_BLE\_DOAview.py module. It gives the possibility of manually setting the following parameters:

```
\overline{1}Fadc = 16e6 #reference frequency of the model
\bar{2}Fs = Fadc#frequency sampling
   N = 8# number of antennas
\mathfrak{X}\overline{4}M = 1# number of signal to detect
   d = 1/2.5#antennas distances (in lambda)
\leq6
\gamma# BLE 5.1 settings
\, 8
   data_rate = 1e6# data rate (or the symbol rate) in bps
   t_s = 10t_d duration = 2*1e-6
\alpha# slot duration of the IQ sampling ( in sec
   t_iq_ssamp= 2*t_slot_duration#iq sampling alternating slots
10f_i_q_samp=1/t_i_q_samp
                                                       #iq samples in freq
11down\_iq=int(np.roomd(Fs/f\_iq\_ samp))#down sampling iq
12
13 -CTETime = 150*N# CTE time (ref. to each antenna) in microsec
14max\_freq\_offset\_res=0e3#residual frequency offset (rfo)
15
   freq_ofiset_res=np.random.randn(1,1)16 -17\,freq\_offset = max\_freq\_offset\_res*(2*freq\_offset\_res[0]-1)fc=0 + freq_offset# Carrier frequency of CTE
18
                                                              #Carrier frequency of CTE
19
20 -21
22
23# URA settings
                                              # SNR of the AWGN added (refered to ADC) in dB
24
   SNR = 60a_{p}hi = np. array ([70, 18, 155]);
                                              # azimuthal angles of arrival under test (dimension
25
         1xM)a_{\text{t}} theta = np. array ([45, 75, 75])
                                              #elevation angles of arravial
26
   Pw_dB = np.array([0, -6, -6]);
                                              # Rx useful power (dBm) of the received signal (
27
        dimension 1XM)
28
   Ch_freq = np.array([fc, fc, fc]);
                                             #signals frequency
                                              #1 = BLE 5.1 CTE, 0 generic GFSK signal
29
   a_CTE\_on = np.array([1,1,1])ovs = int(Fadc/data_rate);# upsampling factor
30 -31mod\_index = 0.5;#modulation index of BLE GFSK signal.between
32
       0.45 - 0.55\mathtt{mod\_type}~=~1\,;#GFSK=1 FSK=0 modulation type
33 -gauss\_del = 234#expected gaussian delay
   apx = np.array([0, 1, 2, 2, 2, 1, 0, 0]) #array of position index on x direction
35 -\frac{1}{2} apy = np. array ([0, 0, 0, 1, 2, 2, 2, 1]) #array of position index on y direction
36
37
   SNR_dB = SNR#Signal Noise to ratio (in dB)
   noise_{energy} = 10 * * ((np.max(Pw_dB)-SNR_dB)/10.0)#noise energy
38
```
The e2e\_URA8\_PureTone\_DOAview.py gives the possibility of manually setting the following parameters:

```
M = 1#number of useful impinging signals
\mathbf{1}N = 8#number of antennas
\gammafc = 2.4e9;# Carrier frequency of the Pure tone
\mathcal{R}d = 1/2.5#antennas distances (in lambda)
\overline{4}Fs = fc*32:
                                                                 # frequency sampling
5
   samples = 100;
                                                              #samples at each antenna
6
\taua_{p}hi = np. array ([30, 78, 155]);
                                                   # azimuthal angles of arrival under test (dimension
\bar{x}1xM)\ddot{Q}a_{\text{i}}theta = np. array ([45, 75, 75])
                                                  #elevation angles of arravial
    Pw_dB = np.array([0, -12, -30]);# Rx useful power (dBm) of the received signal
10 -(dimension 1XM)
11Fch = np. array([0, 2e6, 500e3])
                                                                                  #signals frequency
12
   Fsin = fc+Fch;#signals frequency
13
   apx = np.array([0, 1, 2, 2, 2, 1, 0, 0]) #array of position index on x direction<br>apy = np.array([0, 0, 0, 1, 2, 2, 2, 1]) #array of position index on y direction
14 -15
                                   #Signal Noise to ratio (in dB)
    SNR_dB = 6016
   noise_{energy} = 10 * * ((np.max(Pw_dB) - SNR_dB) / 10.0) #noise energy
17 -t = np.arange(0, samples)*1.0/Fs;# Time
18
```


**Fig. 2.** MUSIC pseudo-spectrum plotted by the DoA view modules.

An example of the plots created by the modules is shown in Fig. 2.

# **Ethics Statement**

The authors declare that the data presented in this article did not involve any use of human subjects, animal experiments nor data collected from social media platforms.

#### **CRediT Author Statement**

**Nicolò Ivan Piazzese:** Conceptualization, Methodology, Software, Formal analysis, Investigation, Writing – original draft, Writing – review & editing; **Michele Perrone:** Conceptualization, Validation, Data curation, Writing – original draft, Writing – review & editing; **Danilo Pietro Pau:** Conceptualization, Validation, Writing – original draft, Writing – review & editing, Supervision, Project administration.

### <span id="page-16-0"></span>**Declaration of Competing Interest**

The authors declare that they have no known competing financial interests or personal relationships which have, or could be perceived to have, influenced the work reported in this article.

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