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Novel Semi-parametric Algorithm for Interference-immune Tunable Absorption Spectroscopy Gas Sensing

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Abstract: One of the most common limits to gas sensor performance is the presence of unwanted interference fringes or etalons arising, for example, from multiple reflections between surfaces in the optical path. Additionally, since the amplitude and the frequency of these interference depend on the distance and alignment of the optical elements, they are affected by temperature changes and mechanical disturbances, giving rise to a drift of the signal. In this work, we present a novel semi-parametric algorithm which allows the extraction of a signal, like the spectroscopic absorption line of a gas molecule, from a background containing arbitrary disturbances, without having to make any assumption on the functional form of these disturbances. The algorithm is applied first to simulated data and then to oxygen absorption measurements in presence of strong fringes. To the best of the authors’ knowledge, the algorithm enables an unprecedented accuracy particularly if the fringes have a free spectral range and amplitude comparable to those of the signal to be detected. The described method presents the advantage of being based purely on post processing, and to be of extremely straightforward implementation if the functional form of the Fourier transform of the signal is known. Therefore it has the potential to enable interference-immune absorption spectroscopy. Finally, its relevance goes beyond absorption spectroscopy for gas sensing since it can be applied to any kind of spectroscopic data.

Keywords: interference; interference cancellation; noise reduction; digital filtering; spectroscopy; sensors.

1. Introduction

Due to the enormous progress in availability and performance of laser light sources and electro-optical components, tunable diode laser absorption spectroscopy (TDLAS) has entered various disciplines both in research and industrial applications. Being a high-sensitivity, selective, fast, non-destructive and in-situ method, TDLAS is currently more and more used for quantitative assessments of gas concentration in several fields. These include, to mention only few, atmospheric environmental monitoring [1–6], medical diagnostics [7–9] chemical analysis [10] and industrial process control [11–13]. The increasing number of applications has pushed the requirements on this method both in terms of sensitivity and in terms of stability. On the other hand, for practical and commercial applications there is a growing interest in compact, simple in design and cost-effective sensitive sensors, which do not require special optical components but guarantee the sensitivity achievable with a complex laboratory equipment.
One of the most common limits to the sensor performance is the presence of unwanted interference fringes called etalons [14]. These interferences may arise due to multiple reflections from reflecting or scattering surfaces in the system, like mirrors, lenses, optical fiber end faces, laser-head windows or dust particles in the gas [14]. Even diffusive surfaces, as for example due to dust deposited on the windows, can give rise to fringes over time [15]. Particularly fringes which have a free spectral range (FSR) of the order of the width of the absorption lines contribute to significant errors in the determination of the line features and require special strategies [16]. Additionally, since the amplitude and the frequency of these interferences depend on the distance and alignment of the optical elements, they are affected by temperature changes and by mechanical disturbances, giving rise for example to a drift of the output signal, thus worsening the long-term performance of the system [17].

The simplest strategy to reduce the effects of interference fringes consists in using anti-reflection coating and wedging or angling of the optical surfaces. Other approaches include, to mention only few, to dither one of the surface creating the interference and integrating the signal so to average out its influence [18,19], the selection of a particular modulation frequency [20] or modulation scheme [21,22], to specifically choose the distance between the interfering surfaces [23] and post-processing filtering [24]. A comprehensive review on signal enhancement and noise reduction techniques can be found in [25]. Although all these approaches have been successfully implemented in the past years, they nonetheless have limitations in the practical implementation depending on the application-specific conditions. For example, it may not always be possible to dither one surface, and the application of a specific detection scheme may limit the flexibility of the measurement.

In this work a new and widely usable approach is presented, which relies only on post processing of the data. Therefore it requires no modification to the apparatus setup or hardware, and can be easily adapted to different experimental configurations. The new presented algorithm allows the extraction of a signal, as the absorption lines of a gas molecule, from a background containing arbitrary disturbances without having to make any assumption on the characteristic of these disturbances in terms of functional form. Therefore, it has the potential to improve the sensitivity and the stability of TDLAS. Furthermore, this algorithm, which is particularly easy in the implementation if the Fourier transform of the signal can be written in closed form, is very general and can be applied to any kind of spectroscopic data. The paper is organized as follows: section 2 describes the fundamentals of the method and of the algorithm; section 3 demonstrates its application to simulated signals; section 4 shows the results for a case of direct absorption spectroscopy for oxygen gas sensing.

2. Description of the algorithm

The algorithm described in this work has the ability to identify a spectroscopic feature from an arbitrary background which needs not to be modeled. The total signal detected in an experiment, here referred to as "total signal" \( I_{tot}(x) \), is modeled as a sum of two contributions: one spectroscopic feature, like an absorption line, here referred to as "signal" \( I(x) \) and a background here referred to as "background" \( B(x) \)

\[
I_{tot}(x) = I(x) + B(x)
\]  

(1)

In the case of direct absorption spectroscopy, \( I(x) \) is the absorbance and \( I_{tot}(x) \) is the distorted absorbance due to \( B(x) \). As mentioned above, the method shines if the background \( B(x) \) cannot be modeled by a known analytical expression. In facts, if \( B(x) \) is not known, is not possible to perform a non-linear fit of \( I_{tot}(x) \) and extract the signal \( I(x) \) without making assumptions on the functional form of \( B(x) \). On the other hand, if the functional form is known but very complex, the algorithm may be advantageous because the inclusion of the background in the non-linear fit may not be possible. Another significant advantage of the proposed algorithm is that the extraction works equally well independently of the amplitude of the interferences, as it will be shown in section 3 and 4.
2.1. Steps of the algorithm: general description

Before describing the steps of the algorithm, the nomenclature and hypothesis for its applicability are introduced. The algorithm is based on the main hypothesis that the Fourier transform of the background $B(x)$ is significantly different than zero only for values of $k$ smaller than a certain cut-off $k_0$

$$F(I_{\text{tot}})(k) = \begin{cases} 
F(I)(k) + F(B)(k) & \text{for } k < k_0 \\
F(I)(k) + c & \text{for } k > k_0
\end{cases} \quad (2)$$

where $F(\cdot)$ denotes the continuous Fourier transform (CFT), $k_0$ a cut-off frequency and $c$ contains the contribution of $F(B)(k)$ for $k > k_0$, which is assumed to be negligible. One central aspect is the determination of a reasonable estimate for this cut-off frequency, as described in the section "Determine the cut-off frequency".

Note that this formulation applies to continuous functions. Since in practice there will always be only a discrete set of points, it is necessary to approximate the CFT in equation (2) by a modified discrete Fourier transform (DFT)

$$D_i(I_{\text{tot}}) = \begin{cases} 
D_i(I) + D_i(B) & \text{for } i < i_0 \\
D_i(I) + c & \text{for } i > i_0
\end{cases} \quad (3)$$

where $D_i(\cdot)$ denotes the modified DFT defined in Appendix A equation A7, $i_0$ the cut-off point corresponding to the cut-off frequency $k_0$, and $c$ contains the contribution of $D_i(B)$ for $i > i_0$, which is assumed to be negligible.

The schematic flow diagram of the algorithm is shown in Fig. 1 to give the reader a high-level understanding of the idea behind it. The single steps are described in detail below.

![Schematic flow diagram of the steps of the algorithm to extract a signal $I(x)$ from a total signal $I_{\text{tot}}(x)$.](image)

**Figure 1.** Schematic flow diagram of the steps of the algorithm to extract a signal $I(x)$ from a total signal $I_{\text{tot}}(x)$.

**Compensate windowing**

In all real experiments the data always cover a limited range in the $x$ direction. For example, the data shown in this paper are measured for a finite laser wavelength range. Mathematically, that is...
equivalent to applying a rectangular window (RW) before calculating the Fourier transform. This is a problem that, if not addressed, will limit the precision that can be achieve with the described algorithm.

In facts, the RW results into making $\epsilon$ in equation 2 not negligible anymore, and thus leads to an error in the fitting procedure of $|F(I(x))(k)|$. This is due to the fact that the DFT of a RW has an amplitude envelope which is proportional to $1/k$, and so it does not goes to zero fast enough [26].

To reduce the effect of windowing considerably the proposed algorithm applies a more intelligent window. The not so often used Tukey window [27,28] has remarkable properties that help tremendously in reducing $\epsilon$ dramatically. The Tukey window, indicated with $T(x)$, is a perfectly flat (constant) symmetric function in the middle that then decrease rapidly to zero on the sides.

The width of the constant part of the Tukey function has to be chosen intelligently. Fig. 2 shows a Lorentzian function with a half width at half maximum (HWHM) indicated as $P_2$ and Tukey function with a width indicated as $W$, both functions normalized to 1 for clarity. As it is easy to understand from Fig. 2, if $W$ is significantly bigger than $P_2$ then $T(x)I(x)$ can be approximated with $I(x)$. Therefore, defining $I_{tot}(x) = T(x)I(x) + T(x)B(x)$, it follows

$$I_{tot}(x) = T(x)I(x) + T(x)B(x)$$  \(4\)

This is the function of which the DFT has to be calculated, instead of simply using $I_{tot}(x)$. Equation 4 can then be approximated under the assumption of $W$ being significantly bigger than $P_2$ as

$$I_{tot}(x) = I(x) + T(x)B(x)$$  \(5\)

$I_{tot}(x)$ is thus the sum of the signal of $I(x)$ and a background which is the product of the original background $B(x)$ and $T(x)$. The modified background $T(x)B(x)$ has a Fourier transform that goes to zero much more rapidly ([27,28]) and that means that $\epsilon$ is much smaller if one considers $I_{tot}(x)$ instead of $I_{tot}(x)$. In other words, while using $I_{tot}(x)$, $\epsilon$ will contain the contribution of $D_i(T \cdot B)$ that is considerably smaller than $D_i(B)$ multiplied by a RW.

Analyzing the deviation between the output of the algorithm and the input signal $I(x)$ with simulated data was performed, it was established that the algorithm works well if the width of the Tukey window is $W \geq 20P_2$. In this work $W = 20P_2$ was used for both the simulated and for the experimental data. The case with $W = 20P_2$ is shown schematically in Fig. 2.

![Figure 2. Schematic representation on how to choose the width of the Tukey window $W$ compared to the Lorentzian HWHM $P_2$. Here is $W = 20P_2$.](image)

**Calculate the DFT**

The step after compensating for the windowing is the calculation of the DFT. To be able to extract directly from the DFT the parameters of the function $I(x)$, it is essential to approximate the CFT by a
modified DFT as described in Appendix A. For all the data shown in this paper the DFT was calculated using the formula (A7).

Determine cut-off

As shown in Fig. 1, the next step is to determine the optimal cut-off point $i_0$ which plays an important role and needs to be chosen carefully. The approach proposed in this work is to choose $i_0$ so to maximize the coefficient of determination $R^2$ obtained by fitting the DFT for $i > i_0$. An example of the implementation is described in the following algorithm written in pseudo code, where $N$ indicates the number of points of the dataset (i.e. the finite number of experimental points), $D$ the DFT of the signal $\tilde{I}_{tot}(x)$, $R_{squared}$ the coefficient of determination $R^2$ calculated from the fit routines, and $R_{limit}$ the value which should be reached for $R^2$. This limiting value can be helpful since beyond a certain value, even if $R^2$ continues to increase when increasing $i_0$, the quality of the fit will not improve significantly and it is not necessary to exclude more points from the DFT for the fit.

```plaintext
pointtoremove = 0
Rsqmin = 0
for i = 0 to N
    remove i points from the data set
    fit the remaining points of D and save the fit parameter Rsquared
    if (i = 0) then pointtoremove = 0
    else if (Rsquared > Rsqmin) then pointtoremove = i and Rsqmin = Rsquared
    if (Rsqmin > Rlimit) then break loop
end of for loop
```

At the end of this loop in the above mentioned pseudo-code $i_0$ is determined and saved in the variable pointtoremove. $R_{squared}$ can be chosen depending on the application. For the curves shown in this paper the loop was stopped for $R_{limit} = 0.99999$. An example of the evolution of $R^2$ with increasing $i_0$ is shown in Fig. 3. The data refer to the third scenario described later in section 3. Above $i_0 = 30$, $R^2$ still continue to slightly increase but the statistical goodness of the fit does not improve further.

![Figure 3](Image)

**Figure 3.** Evolution of $R^2$ with increasing value of the cut-off $i_0$. The data correspond to the third scenario described in section 3.

Fit of DFT

The final step of the algorithm is to perform a non-linear fit of the DFT for $i > i_0$. Since the functional form of the DFT is known this is a standard procedure which can be performed, for example, with least-square-fit routines and will not be discussed here. Since the functional form of $|F(I(x))(k)|$ is known, after the fit $I(x)$ is determined without the need of doing an inverse Fourier transform.
2.2. Algorithm applied to a Lorentzian line shape

In this section the implementation of the algorithm to a Lorentzian function signal $I(x)$ is described as an example. This functional form was chosen because describes the absorption lines of many gas molecules, as for example oxygen under atmospheric conditions. In other conditions, like at higher temperatures or lower pressures, the Gaussian contribution due to Doppler broadening cannot be neglected and a Voigt profile is a better description.

The Lorentzian function can be written as

$$I(x) = \frac{P_1 P_2}{\pi(x^2 + P_2^2)}$$

with $P_1, P_2 > 0$. In this form $P_1$ and $P_2$ represent the area and the HWHM of the line. Writing $I(x)$ in this form is particularly advantageous since in direct absorption spectroscopy the gas concentration can be determined directly from the area under the line, thus is given directly by $P_1$. In this formulation $|F(I(x))(k)|$ is then a simple exponential

$$F(I(x))(k) = P_1 e^{-P_2 |k|}$$

So, once the parameters $P_1$ and $P_2$ are determined from the fit of the DFT, $I(x)$ is known.

3. Application to simulated data

The novel algorithm was first applied to artificially simulated data to demonstrate its functioning and its performance. Since the signal to be extracted $I(x)$ is known, it is possible to estimate the accuracy and robustness of the algorithm in presence of backgrounds with different characteristics. In particular, three scenarios with different types of periodic disturbances were simulated. The signal to be extracted $I(x)$ is for all three cases the same Lorentzian function written in the form of Eq. 6 with $P_1 = 5\pi$ and $P_2 = 5$. All three scenarios were chosen to reflect real cases which are typical of TDLAS.

The first scenario is chosen to represent the experimental situation when the background has a periodic disturbance with a FSR comparable to the width of the line to be detected. This type of background is particularly problematic because it strongly affects the determination of the line shape. Furthermore, it cannot be removed by introducing a small jitter on the laser diode laser current and averaged out [29] or be filtered out with standard post-processing methods, without introducing a significant distortion to the line shape. This background taken here is a simple cosine function

$$B(x) = 0.07 \cos(0.1x + 1)$$

The total signal $I_{tot}(x)$ in this case, together with the expected signal $I(x)$ are shown Fig. 4 on the left. Also shown in the figure on the left is the result obtained by applying the described algorithm. Despite the problematic background, the output of the algorithm is practically identical to $I(x)$. The percent deviation of the two parameters $P_1$ and $P_2$ describing the Lorentzian obtained with the algorithm from the initial value used to generate $I(x)$ is only of 0.27% for $P_1$ and 0.23% for $P_2$.

To better illustrate the contribution of the background, the DFT of the Lorentzian and the DFT of the total signal signal are also shown in Fig. 4 on the right. The two peaks visible in the figure represent the contribution of the background. With the pseudo-code algorithm described in section 2.1 the cut-off was $i_0 = 13$.

The second scenario discussed here is that of a background with a weak disturbance characterized by a FSR much larger than the line width and as large as or larger than the measuring range. This type of disturbance arises because of reflections between two surfaces separated by a very short physical
Figure 4. First scenario: disturbance with a FSR comparable to the line width. Left: Simulated experimental total signal $I_{\text{tot}}(x)$ (blue line), Lorentzian line shape signal $I(x)$ (black dots) and extracted signal obtained with the algorithm (red line). Right: DFT of the total signal $|F(I_{\text{tot}}(x))(k)|$ (red line) and DFT of the Lorentzian signal $|F(I(x))(k)|$ (black points).

The result of the algorithm for this scenario are shown in Fig. 5: on the left are plotted the total signal $I_{\text{tot}}(x)$, the expected signal $I(x)$ and the result obtained by applying the algorithm. In Fig. 5 on the right are shown the DFT of the Lorentzian and the DFT of the total signal signal. The two peaks due to the contribution of the background are now very close to zero, which makes the extraction particularly unproblematic. With the pseudo-code algorithm described in section 2.1 the cut-off was $i_0 = 7$.

Figure 5. Second scenario: weak disturbance with a FSR as large as the measuring window. Left: Simulated experimental total signal $I_{\text{tot}}(x)$ (blue line), Lorentzian line shape signal $I(x)$ (black dots) and extracted signal obtained with the algorithm (red line). Right: DFT of the total signal $|F(I_{\text{tot}}(x))(k)|$ (red line) and DFT of the Lorentzian signal $|F(I(x))(k)|$ (black points).

Also in this case it is clear from the figure that the algorithm extracts exceedingly well the signal. The percent deviation of the two parameters describing the Lorentzian obtained with the algorithm is only of 0.28% for $P_1$ and 0.25% for $P_2$.

As a third scenario, a background function resulting from the sum of a hundred cosine functions is considered. This less realistic case is chosen to demonstrate that, no matter how dramatic the
interferences are, the algorithm can extract signal $I(x)$ very well. Also, this scenario illustrate the case when the functional form of the background is too complex to be included in a non-linear fit of $I_{\text{tot}}(x)$. The background is thus written as

$$B(x) = \sum_{i=1}^{100} A_i \cos(w_i x + \phi_i) \quad (10)$$

where $w_i$ are chosen randomly from a normal distribution with average equal to zero and a standard deviation of 0.1, $\phi_i$ from a normal distribution with average equal to zero and a standard deviation of 0.2, and $A_i$ from a normal distribution with average equal to zero and a standard deviation of 0.03.

In Fig. 6 on the left are shown the total signal $I_{\text{tot}}(x)$, the expected signal $I(x)$ and the result obtained by applying the described algorithm. Despite the very complicated background the extraction of the signal $I(x)$ by the algorithm works very well. In Fig. 6 on the right the DFT of the Lorentzian and the DFT of the total signal signal are also shown. Due to the high number of cosine functions in the background the DFT has a much structured shape. With the pseudo-code algorithm described in section 2.1 the cut-off was $i_0 = 30$ (see also Fig. 3).

![Figure 6](image_url)

**Figure 6.** Third scenario: strong multiple disturbances. Left: Simulated experimental total signal $I_{\text{tot}}(x)$ (blue line), Lorentzian line shape signal $I(x)$ (black dots) and extracted signal obtained with the algorithm (red line). Right: DFT of the total signal $|F(I_{\text{tot}}(x))(k)|$ (red line) and DFT of the Lorentzian signal $|F(I(x))(k)|$ (black points).

The percent deviation of the two parameters describing the Lorentzian obtained with the algorithm for this scenario is only of 0.012% for $P_1$ and 0.14% for $P_2$. The deviation of both parameters is particularly low in this case. To better estimate the error and the standard deviation on the parameters the method was applied to 500 functions created with the random sum of 100 cosines described above. Then the error was evaluated and its distribution studied. As a result $P_1$ has a mean value of the absolute value of the percentual error of 0.12% with a standard deviation of 0.19%, and $P_2$ a mean of 0.04% with a standard deviation of 0.06%.

Finally, the analysis of the algorithm with simulated data has allowed to determine the causes of the discrepancy between the parameters extracted with the algorithm and the starting function $I(x)$. The main contribution to these arises from the approximation of the CFT by a DFT and is due to a rather large point spacing and the limited x-range of $I(x)$ used in the simulated data. Smaller contributions arise from an imperfect window compensation, and a very small frequency folding [30], which was neglected here. Since the purpose of this paper is to illustrate the algorithm and not to minimize the discrepancies, the simulated data were chosen to be as close as possible to typical experimental data. The application to the three scenarios demonstrate well how, even with a very complicated background like in the third one, the proposed algorithm can extract the underlying signal extremely well.
4. Experimental Results

To demonstrate the robustness of the method on real gas sensing measurements, absorption spectroscopy was performed on the three strong lines R9R9 (760.77 nm), R7Q8 (760.89 nm) and R7R7 (761.00 nm) of the O₂ near infrared A-band in presence of multiple interference fringes.

4.1. Experimental setup

The setup for the absorption spectroscopy experiments was chosen to be extremely simple and is shown schematically in Fig. 7.

![Figure 7. Schematic diagram of the setup used for the absorption spectroscopy experiments.](image)

The light source is a 0.25 mW single-mode VCSEL (760 nm TO5 VCSEL, Philips Photonics) emitting at 760 nm. The laser current and temperature were adjusted by a temperature controller (TEC 2000, Thorlabs) and a VCSEL laser diode controller (LDC 200C, Thorlabs). The light emitted by the laser is collimated by a lens. The light transmitted by the sample is collected using a large-area Si 10x10 mm² photodiode (FS1010, Thorlabs), amplified by an adjustable-gain photodiode amplifier (PDA200C, Thorlabs). The current ramp for the wavelength-sweep and the data acquisition were performed by a DAQ card (USB-6361, National Instrument) using a Labview™ software. The laser current sweep was chosen so to be able to measure three oxygen absorption lines. The total distance between the laser and the detector was kept fixed at approximately 36 cm. Interference fringes of adjustable intensity were generated by inserting and tilting a window of known material in the optical path. By varying the thickness of the window it is possible to achieve fringes with different FSR; by tilting the window it is possible to adjust the amplitude of the fringes. In this work two windows of BK7 of thicknesses d=11 mm (window 1) and d=4 mm (window 2) were used. These two windows were chosen to create interferences fringes with FSR comparable to (window 1) and greater than (window 2) the line width of the signal. This type of interferences are the most disturbing because they cannot be easily eliminated, for example by a small jitter in the laser current or by standard post-processing filtering.

4.2. Oxygen sensing

The absorbance signal of the three oxygen lines R9R9, R7Q8 and R7R7 is shown in Fig. 8. The interference fringes due to the window are clearly visible. Superimposed to these fringes, other minor ones are also visible, which are due to the glass window of the laser package and to the surfaces of the collimator. The zero-absorbance baseline arising from the non-linear wavelength-dependent intensity of the laser was determined by a fourth-order polynomial fit. All the measurements reported in this work were performed in air, at room temperature and ambient pressure.

Both measurements shown in Fig. 8 were processed with the proposed algorithm assuming a Lorentzian line shape. The results for the line R7Q8 is shown in Fig. 9. For comparison also the expected absorption lines based on the HITRAN 2012 database [31] were carried out using the application SpectraPlot [32]. It is evident from Fig. 9, and particularly from the enlargement on the
right, that the algorithm extracts the absorption line extremely well. Differently from many fitting methods, the algorithm does not require neither input parameters nor initial values. The slightly lower peak height for the line in presence of the window 1 is due to the fact that the distance between laser and detector was kept fixed and the window 1 is thicker, thus reducing the optical path length for oxygen absorption of \((11 - 4) \, \text{mm} = 7 \, \text{mm}\). This difference in the HITRAN simulated curves was extracted correctly by the algorithm. To estimate the accuracy of the algorithm, the area under the absorption line from the HITRAN database calculated numerically was compared with the value of \(P_1\) extracted with the algorithm. For both measurements the difference is 0.1\%. There are still minimal deviations between the extracted and simulated lines, which could be reduced by considering a Voigt instead of a purely Lorentzian profile for the oxygen lines.

5. Conclusions

In this work, a novel semi-parametric algorithm is presented, which allows the extraction of a signal from an arbitrary background. In particular, the algorithm is applied to a background which is the sum of periodic interference fringes of different amplitude and FSR. These type of disturbances,
arising for example from multiple reflections between surfaces in the optical path, are of highest relevance for absorption spectroscopy because they are frequently the most common limit to gas sensor performance. The novel algorithm is first demonstrated on simulated data for three scenarios chosen to represent particularly relevant practical situations. In all the three cases, the discrepancy between the results obtained with the algorithm and the expected values for the line parameters is very small, less 0.3%. Then, the algorithm is applied to experimental data of oxygen absorption in presence of multiple interference. Despite the strong fringes, the extracted line shows a remarkable agreement with the expected curves from the HITRAN database, with deviation of the area of only 0.1%. These results show that the performance of a very simple sensing setup, with poor anti-reflection coatings and minimal precautions to minimize the interference fringes, can be strongly improved by simply post processing the data with the proposed algorithm.

In this paper the algorithm was applied to a Lorentzian line shape in direct absorption spectroscopy for oxygen concentration determination. However, it can be applied to account for other line shapes, for example a Voigt profile, or to signals with a different functional form, as arising for example from wavelength-modulation spectroscopy. The main advantage of the method is that no starting or input parameters are necessary. Particularly in the case of fringes with large amplitudes and FSR of the order of the FWHM of the line, it can extract the desired signal very well, making the signal insensitive to fringes and their change with time.

In conclusion, the presented algorithm, being able to extract the signal feature from an arbitrary background has the potential to allow interference-immune TDLAS, solving long-time stability problems arising from changes over time of the background, like thermal drift. Furthermore, this algorithm is not specific of TDLAS and can be applied to any kind of spectroscopic data provided the functional shape of the signal to be detected is known.

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Author Contributions: F.V. conceived, designed and performed the experiments; U.M. proposed and developed the algorithm; F.V. and U.M. analyzed the data and wrote the paper.

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Abbreviations

The following abbreviations are used in this manuscript:

- **TDLAS**: tunable diode laser absorption spectroscopy
- **FSR**: free spectral range
- **CFT**: continuous Fourier transform
- **DFT**: discrete Fourier transform
- **RW**: rectangular window
- **FWHM**: half width at half maximum

Appendix A  Approximation of the CFT by the modified DFT

In this paper the following definitions for the continuous Fourier transform (CFT)

\[ F(k) = \int_{-\infty}^{\infty} f(x) e^{-ixk} dx \]  \hspace{1cm} (A1)

and for the discrete Fourier transform (DFT)

\[ D_n(f) = \sum_{j=0}^{m-1} f(x_j) e^{-2\pi ij n / m} \] \hspace{1cm} (A2)
are used, where \( m \) indicates the number of data points at disposal. For convenience \( m \) is taken to be even.

Let’s first make some assumptions that will simplify the calculations. Let’s assume that the function \( f(x) \) is zero outside a certain range \((-a/2, a/2)\) with some \( a \in \mathbb{R} \) and \( a > 0 \). Let’s introduce the sampling rate \( \beta \) as

\[
\beta = \frac{a}{m} \quad (A3)
\]

The abscissas of the data \( x_j \) and \( k_j \) can be written as:

\[
x_j = \left(j - \frac{m}{2}\right) \beta \quad (A4)
\]

\[
k_j = \frac{2\pi}{a} \left(j - \frac{m}{2}\right) \quad (A5)
\]

with \( j \) going from 0 to \( m \). We can approximate equation (A1) with a Riemann sum using the fact that \( f(x) \) is zero outside the range \((-a/2, a/2)\):

\[
F(k_n) = \int_{-\infty}^{\infty} f(x) e^{-ixk_n} dx = \int_{-a/2}^{a/2} f(x) e^{-ixk_n} dx \implies F(k_n) = \beta \sum_{j=0}^{m} f(x_j) e^{-ixk_n} \quad (A6)
\]

where we have approximated the integral with a discrete sum. Clearly, the bigger \( m \) is, the better will be the approximation. Now using equations (A4) and (A5) we can rewrite (A6) as

\[
F(k_n) = \beta \sum_{j=0}^{m} f(x_j) e^{-2\pi i (j - m/2)(n - m/2)/m} \quad (A7)
\]

note that the exponent in (A7) can be rewritten as

\[
-2\pi i \left(\frac{jm}{m} - \frac{nm}{2} + \frac{m^2}{4}\right) \frac{1}{m} = -2\pi i \left(\frac{jm}{m} + \pi ij + \pi in - \pi i \frac{m}{2}\right) = \pi i \left(n - \frac{m}{2}\right) + \pi ij - 2\pi i \frac{jm}{m} \quad (A8)
\]

so from (A7) and (A8) it follows that the CFT calculated in \( k_n \) can be approximated as

\[
F(k_n) = \beta e^{\pi i (n - m/2)} \sum_{j=0}^{m} f(x_j) e^{\pi ij} e^{-2\pi ijn/m} = (-1)^n \beta \sum_{j=0}^{m-1} f(x_j) (-1)^j e^{-2\pi ijn/m} \quad (A9)
\]

where with \( \{(-1)^j f(x_j)\} \) we have indicated the set of datapoints that we have at our disposal to calculate the DFT. \( D_n \) in this equation is calculated from equation (A9) with \( f(x_j) \) multiplied by the factor \((-1)^j\).
References


