Fast minimum variance wavefront reconstruction for extremely large telescopes

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We present a new algorithm, FRiM (FRactal Iterative Method), aiming at the reconstruction of the optical wavefront from measurements provided by a wavefront sensor. As our application is adaptive optics on extremely large telescopes, our algorithm was designed with speed and best quality in mind. The latter is achieved thanks to a regularization which enforces prior statistics. To solve the regularized problem, we use the conjugate gradient method which takes advantage of the sparsity of the wavefront sensor model matrix and avoids the storage and inversion of a huge matrix. The prior covariance matrix is however non-sparse and we derive a fractal approximation to the Karhunen-Loève basis thanks to which the regularization by Kolmogorov statistics can be computed in $O(N)$ operations, $N$ being the number of phase samples to estimate. Finally, we propose an effective preconditioning which also scales as $O(N)$ and yields the solution in 5–10 conjugate gradient iterations for any $N$. The resulting algorithm is therefore $O(N)$. As an example, for a $128 \times 128$ Shack-Hartmann wavefront sensor, FRiM appears to be more than 100 times faster than the classical vector-matrix multiplication method.

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1. Introduction

The standard and most used method for adaptive optics (AO) control is based on a vector-matrix multiply (VMM) of the vector of wavefront sensor measurements by the so-called control matrix [1]. This operation gives an update of the commands to be sent to the deformable mirrors to adjust the correction of the corrugated incoming wavefronts. The control matrix is precomputed, generally using modal control optimization [2]. The complexity of computing the control matrix using standard methods scales as $O(N^4)$, where $N$ is the number of unknowns (phase samples or actuator commands), and applying real time VMM scales as $O(N^2)$. This computational burden can be reasonably handled on current AO systems where $N \leq 10^3$.

For future Extremely Large Telescopes (ELT’s), the number of actuators being considered is in the range $10^4$ – $10^5$. This huge increase is the result of both the larger diameter of the ELTs [3] and the emergence of new architectures for the AO systems, using either a greater density of actuators (Extreme AO) or combining several deformable mirrors and wavefront sensors (multi-conjugate AO, multi-object AO) [4]. The necessary computational power for real time control on such systems is currently unattainable when using standard methods.

More efficient algorithms are thus required and have been developed in recent years. Poyneer et al. [5] have derived an accurate Fourier transform wavefront reconstructor by solving the boundary problem in circular apertures. This reconstructors scale as $O(N \log N)$ and is shown to be effective for Extreme AO [6]. MacMartin [7] studied several approximate approaches such as a multiple-layer hierarchic reconstruction, which scales as $O(N)$.

Although least-squares algorithms give suitable results for single star AO systems (classical on-axis AO or Extreme AO), minimum variance reconstruction is required to minimize the effects of the missing data or unseen modes in the other AO schemes [8]. In the context of minimum variance for multi-conjugate AO, Ellerbroek [9] could apply sparse matrix techniques (Cholesky factorization) using a sparse approximation of the turbulence statistics, and introducing as low-rank adjustments, the nonsparse matrix terms arising from the global tip/tilt measurement errors associated with laser guide stars. However the interactions between the layers in their tomographic modeling reduce the efficiency of the sparse direct decomposition methods [10].

Iterative methods are also extensively studied in this context. Their main asset is their ability to iteratively compute the unknowns from the measurements using direct sparse matrices, and so the storage of a precomputed inverse full matrix is not necessary. One major problem with iterative methods is the increase in the number of iterations with the number of unknowns to estimate [11–13]. As an example, Wild et al. [14] have proposed to use the closed-loop AO system itself as an iterative processor, but the performance of the least squares reconstruction depends on the loop frequency of the AO system, which should be higher than usual.

The most successful iterative methods in AO are now based on preconditioned conjugate gradients (PCG) [15], where...
some of the previous approximate reconstruction methods are embedded as preconditioners to ensure a small number of iterations (see section 4). Gilles et al. [16] have described a multigrid PCG algorithm, mainly aimed at Extreme AO and scaling as $O(N \log N)$. The multigrid preconditioner is somewhat related to the multiple-layers hierarchic reconstruction [7]. This wavefront reconstruction method has been improved with a faster approximation to the turbulence statistics, scaling as $O(N)$ [17]. The multigrid PCG algorithm has also been developed for multi-conjugate AO [18]. In this case, the structure of the matrix is more complex and brings some limitations. More recently, a Fourier domain preconditioner was introduced [19, 20] in the context of multi-conjugate AO, with a faster reconstruction than multigrid PCG. In this case, the preconditioner is related to the Fourier transform wavefront reconstructor [5]. Both multigrid and Fourier domain preconditioners were examined for the Thirty Meter Telescope project [21, 22].

In this work, we propose novel methods to address the two critical points previously seen in iterative methods for wavefront reconstruction: estimation of the atmospheric phase covariance matrix and preconditioning.

We need a sparse representation of the inverse of the atmospheric phase covariance matrix to efficiently introduce priors in the minimum variance estimator. Currently, we can choose between a good representation in the Fourier domain with $O(N \log N)$ complexity [16, 19] and a widely used sparse biharmonic approximation introduced by Ellerbroek [9], less accurate [19], but scaling as $O(N)$. With FRiM, we introduce a so-called “fractal operator” as a multiscale algorithm with $O(N)$ complexity. This operator, both accurate and very fast, was inspired by the mid-point method of Lane et al. [23] to generate a Kolmogorov phase screen. It can be used for any wavefront structure function. It allows us to very efficiently apply the inverse of the phase covariance matrix to any vector.

We show that this fractal operator is also very efficient when used as a preconditioner. It allows the wavefront reconstruction to be iteratively computed in a space of statistically independent modes. We additionally use a classical Jacobi preconditioner, or a new “optimal diagonal preconditioner” to further improve the convergence. The number of iterations is $\leq 10$ for a full wavefront reconstruction whatever the size of the system, with a number of floating point operations $\sim 34 \times N$ per iteration. The method is therefore globally $O(N)$.

In the following, we first derive the analytical expression for the minimum variance restored wavefront and the equations to be solved. We then introduce the fractal operator allowing fast computation of the regularization term in an iterative method such as conjugate gradients. We then propose two fast preconditioners to further speed up the iterative algorithm. We finally use numerical simulations to test the performances of FRiM.

2. Minimum variance solution

A. Model of data

We assume that the wavefront sensor provides measurements of spatial derivatives (slopes or curvatures) of the phase, which are linearly related to the wavefront seen by the sensor:

$$d = S \cdot w + n$$

(1)

where $d \in \mathbb{R}^N$ is the data vector provided by the sensor, $w \in \mathbb{R}^N$ is the vector of sampled wavefront values, $S \in \mathbb{R}^{N \times N}$ is the sensor response matrix and $n \in \mathbb{R}^M$ accounts for the noise and model errors. This equation is general as long as the wavefront sensor is linear. As a typical case, we will however consider a Shack-Hartmann wavefront sensor with Fried geometry [24] in our simulations and for the evaluation of the efficiency of the algorithms.

B. Optimal wavefront reconstructor

The estimation of the wavefront $w$ given the data $d$ is an inverse problem which must be solved using proper regularization in order to improve the quality of the solution while avoiding noise amplification or ambiguities due to missing data [25]. In order to keep the problem as simple as possible, we first introduce the requirement that the solution be a linear function of the data, i.e. the restored wavefront satisfies:

$$\tilde{w} = R \cdot d$$

(2)

where $R$ is the restoration matrix and $d$ the wavefront sensor measurements. Some quality criterion is needed to derive the expression for the restoration matrix $R$. For instance, we can require that, on average, the difference between the restored wavefront $\tilde{w}$ and the true wavefront $w$ be as small as possible by minimizing $\langle \|\tilde{w} - w\|^2 \rangle$ where $\langle \cdot \rangle$ denotes the expected value of its argument. It is interesting to note that minimizing (on average) the variance of the residual wavefront yields the optimal Strehl ratio [26] since:

$$SR \approx \exp \left( -\frac{1}{\mathcal{A}} \int_{\text{pupil}} \langle [\tilde{w}(r) - w(r)]^2 \rangle dr \right)$$

(3)

where $r$ is the position in the pupil, $\mathcal{A}$ is the area of the pupil and $w(r)$ is the wavefront phase in radian units. The best reconstruction matrix according to our criterion then satisfies:

$$R^\dagger = \arg \min_{R} \langle \| R \cdot d - w \|^2 \rangle.$$  

(4)

Accounting for the facts that the wavefront $w$ and the errors $n$ are uncorrelated and have zero means, i.e. $\langle n \rangle = 0$ and $\langle w \rangle = 0$, the minimum variance reconstructor expands as [27]:

$$R^\dagger = C_w \cdot S^T \cdot (S \cdot C_w \cdot S^T + C_n)^{-1},$$

(5)

where $C_n \equiv \langle n \cdot n^T \rangle$ is the covariance matrix of the errors and $C_w \equiv \langle w \cdot w^T \rangle$ is the a priori covariance matrix of the wavefront. Applying this reconstructor to the data $d$ requires solving a linear problem with as many equations as there are measurements. Generally, wavefront sensors provide more measurements than wavefront samples (about twice as many for a
Shack-Hartmann or a curvature sensor). Fortunately, from the following obvious identities [28]:
\[
S^T \cdot C_n^{-1} \cdot S \cdot C_w \cdot (S^T + S) = \left(S^T \cdot C_n^{-1} \cdot (S + C_w^{-1}) \right) \cdot C_w \cdot S^T,
\]
we can rewrite the optimal reconstructor in Eq. (5) as:
\[
R^I = \left(S^T \cdot C_n^{-1} \cdot (S + C_w^{-1}) \right)^{-1} \cdot S^T \cdot C_n^{-1}
\] (6)
which involves solving just as many linear equations as there are wavefront samples. The linear reconstructor defined in Eq. (6) is the expression to be preferred in our case.

C. Links with other approaches

Using Eq. (6) for the reconstructor, the minimum variance restored wavefront is given by:
\[
w_I = R^I \cdot d = \left(S^T \cdot C_n^{-1} \cdot (S + C_w^{-1}) \right)^{-1} \cdot S^T \cdot C_n^{-1} \cdot d
\]
which is also the solution of the quadratic problem:
\[
w_I = \arg \min_w \left\{ (S \cdot w - d)^T \cdot C_n^{-1} \cdot (S \cdot w - d) + w^T \cdot C_w^{-1} \cdot w \right\}
\]
where \((S \cdot w - d)^T \cdot C_n^{-1} \cdot (S \cdot w - d)\) is the so-called \(\chi^2\) which measures the discrepancy between the data and their model and \(w^T \cdot C_w^{-1} \cdot w\) is a Tikhonov regularization term which enforces a priori covariance of the unknowns. Thus Eq. (6) is also the result of the maximum a posteriori (MAP) problem. Here, the usual hyper-parameter is hidden in \(C_w\) which is proportional to \((D/r_0)^{5/3}\), where \(r_0\) is the Fried parameter [29]. As already noted by other authors (see e.g. Rouset [30]), the minimum variance estimator is directly related to Wiener optimal filtering.

Actual adaptive optics systems make use of some expansion of the wavefront on a basis of modes, regularizing being achieved by setting the ill-conditioned modes to zero. This technique is similar to truncated singular value decomposition (TSVD) [30]. Since truncation results in aliasing, we expect that the MAP solution will be a better approximation to the wavefront.

D. Iterative Method

The optimal wavefront can be computed in different ways. For instance, the matrix \(R\) can be computed once, using Eq. (5) or Eq. (6), and then applied to every data set \(d\). Since it requires the numerical inversion of an \(N \times N\) matrix, the direct computation of \(R\) scales as \(O(N^3)\) operations [13]. The reconstructor \(R\) is a \(N \times M\) matrix and is not sparse in practice. Hence, the storage of \(R\) requires \(MN \approx 2N^2\) floating point numbers and computing \(R \cdot d\) requires \(\approx 2MN \approx 4N^2\) floating point operations. For large numbers of degrees of freedom \(N \sim (D/r_0)^2\), the computer time spent by the matrix-vector multiplication can be too long for real time applications. Moreover the memory requirement (e.g. for \(N \approx 10^4\), 1.5 Gb of memory are needed to store \(R\)) may be such that memory page faults dominate the computation time of matrix-vector multiplication.

initialisation:
compute \(r_0 = b - A \cdot x_0\) for some initial guess \(x_0\)
let \(k = 0\)
until convergence do
solve \(M \cdot z_k = r_k\) for \(z_k\) (apply preconditioner)
\(p_k \leftarrow r_k\)
if \(k = 0\), then
\(p_k \leftarrow z_k\)
else
\(p_k \leftarrow z_k + (\rho_k/p_{k-1}) p_{k-1}\)
endif
\(q_k = A \cdot p_k\)
\(a_k = \rho_k/(p_k^T \cdot q_k)\)
\(x_{k+1} = x_k + a_k \cdot q_k\)
\(r_{k+1} = r_k - a_k \cdot q_k\)
k \(\leftarrow k + 1\)
done

Fig. 1. Preconditioned conjugate gradient algorithm for solving \(A \cdot x = b\) where \(A\) is a symmetric positive definite matrix and \(M\) is a preconditioner. The unpreconditioned version of the algorithm is simply obtained by taking \(M = I\), hence \(z_k = r_k\).

In order to avoid the direct matrix inversion and the matrix-vector product required by the explicit computation of \(R\), we use an iterative method to solve the linear system
\[
\left(S^T \cdot C_n^{-1} \cdot (S + C_w^{-1}) \right) \cdot w = S^T \cdot C_n^{-1} \cdot d
\] (7)
which leads to the optimal wavefront \(w\) for every data set \(d\). For the purpose of the discussion, Eq. (7) can be put in a more generic form:
\[
A \cdot x = b
\] (8)
where, in the case of Eq. (7), \(x = w\) and:
\[
A = S^T \cdot C_n^{-1} \cdot (S + C_w^{-1})
\] (9)
is the so-called left hand side matrix, whereas
\[
b = S^T \cdot C_n^{-1} \cdot d
\] (10)
is the so-called right hand side vector.

Barrett et al. [15] have reviewed a number of iterative algorithms for solving linear systems like (8). An advantage of these methods is that they do not explicitly require the matrix \(A\); it is sufficient to be able to compute the product of matrix \(A\) (or its transpose) with any given vector. The iterative algorithm therefore fully benefits from the possibility to compute the matrix-vector products in much less than \(O(N^2)\) operations when \(A\) is sparse or has some special structure. This is particularly relevant in our case since applying \(A\) can be achieved by matrix-vector products very fast to compute as shown in Sect. 2E and Sect. 2F. The drawback of iterative methods is that the computational burden scales as the number of iterations required to approximate the solution with sufficient precision. In the worst case, the number of iterations can theoretically be as high as the number of unknowns...
shows the steps of the CG algorithm to solve the system Eq. (7) can result in a much higher number of iterations, even on small systems. This problem can however be greatly reduced by means of a good preconditioner [12, 15].

By construction, $A$ given by Eq. (9) is a symmetric positive definite matrix and the conjugate gradient (CG) [15] is the iterative method of choice to solve the system in Eq. (8).

Figure 1 shows the steps of the CG algorithm to solve the system $A \cdot x = b$. This method is known to have a super-linear rate of convergence [12], and can be accelerated by using a proper preconditioner $M \approx A$ for which solving $M \cdot z = r$ for $z$ (with $r = b - A \cdot x$) is much cheaper than solving Eq. (8) for $x$. The preconditioner can also be directly specified by its inverse $Q = M^{-1}$ such that $Q \approx A^{-1}$ and then $z = Q \cdot r$ in the CG algorithm. Without a preconditioner, taking $M = Q = I$, where $I$ is the identity matrix, yields the unpreconditioned version of the CG algorithm. In Sect. 4 we investigate various means to obtain an effective preconditioner for the wavefront reconstruction problem.

In the remainder of this section, we derive means to quickly compute the dot product with the matrix $A$ in Eq. (9). To that end, we consider separately the Hessian matrix $S^T \cdot C_n^{-1} \cdot S$ of the likelihood term and that of the regularization term $C_w^{-1}$.

E. Computation of the likelihood term

Most adaptive optics systems use either a Shack-Hartmann sensor which provides measurements of the local gradient of the wavefront or a curvature sensor which measures the local curvature of the wavefront [1]. Since such sensors probe local spatial derivatives of the wavefront, their response can be approximated by local finite differences which yields a very sparse linear operator $S$. Though some non-sparse matrix terms can appear due to tilt indetermination with laser guide stars or to take account of natural guide star tip/tilt sensors. Owing to the low rank of these modes, sparse matrix models can still be applied [9]. Thus, denoting $N_{\text{diff}}$ the number of wavefront samples required to compute the local finite differences, only $M \times N_{\text{diff}}$ out of $M \times N$ coefficients of $S$ are non-zero. For instance, Fig. 2 shows the Fried geometry of the Shack-Hartmann sensor model [24] which we used in our numerical simulations. The error free slopes are related to the wavefront by:

$$
\begin{align*}
    d_x(x,y) &= \frac{1}{a} \left[ w(x+a,y+a) + w(x+a,y) \\
    &- w(x,y+a) - w(x,y) \right] \\
    d_y(x,y) &= \frac{1}{a} \left[ w(x+a,y+a) - w(x+a,y) \\
    &+ w(x,y+a) - w(x,y) \right]
\end{align*}
$$

where $(x,y)$ are the pupil coordinates, $d_x$ and $d_y$ are the slopes along the $x$ and $y$ directions and $a$ is the sampling step. Hence $N_{\text{diff}} = 4$, in our case, whatever the number of degrees of freedom. Besides, to a good approximation, wavefront sensors provide uncorrelated measurements [1], hence the covariance matrix $C_n$ of the errors can be taken as a diagonal matrix:

$$
C_n \approx \text{diag}(\text{Var}(n))
$$

where Var$(n)$ is the vector of noise and error variances. Since $C_n$ is diagonal, its inverse $C_n^{-1}$ is diagonal and trivial to compute. Finally, the matrices $S$ and $C_n^{-1}$ are sparse and the dot product by $S^T \cdot C_n^{-1} \cdot S$ can be therefore computed in $O(N)$ operations.

F. Fast estimation of the regularization term

Unlike $C_n$ and $C_n^{-1}$, neither $C_w$ nor $C_w^{-1}$ is sparse. We introduce here a way to derive an approximation for $C_w^{-1}$ so that it can be applied to a vector with a small number of operations. We first consider the following decomposition of $C_w$:

$$
C_w = K \cdot K^T
$$

where $K$ is a square invertible matrix. Since $C_w$ is positive definite, there exists a number of possibilities for such a factorization: Cholesky decomposition [9, 13, 17], spectral factorization, etc. We then use this decomposition to define new variables $u$ based on the wavefront $w$:

$$
u \stackrel{\text{def}}{=} K^{-1} \cdot w.
$$

The expected value of $u$ is: $\langle u \rangle = K^{-1} \cdot \langle w \rangle = 0$ and its covariance matrix therefore satisfies:

$$
C_u = \langle u \cdot u^T \rangle = K^{-1} \cdot \langle w \cdot w^T \rangle \cdot K^T = K^{-1} \cdot C_w \cdot K^T = I,
$$

which shows that the new variables are independent and identically distributed following a normal law: $u \sim N(0, I)$. This gives rise to a method for generating wavefronts since from a set $u$ of $N$ independent random values following a normal law, taking $w = K \cdot u$ yields a random wavefront with the expected covariance. Finally, using this re-parametrization, it is possible to rewrite the regularization term as:

$$
\begin{align*}
    w^T \cdot C_w^{-1} \cdot w &= w^T \cdot K^{-T} \cdot K^{-1} \cdot w \\
    &= \|K^{-1} \cdot w\|_2^2 = \|u\|_2^2.
\end{align*}
$$

Then, depending on whether the problem is solved for the wavefront samples $w$ or for the so-called wavefront generators $u$ (cf. equations (38) and (39) in Sect. 5), each conjugate
the corresponding
3. Fractal operators
A. Principle and structure function
The mid-point algorithm [23] starts at the largest scales of the wavefront and step-by-step builds smaller scales by interpolating the wavefront values at the previous scale and by adding a random value with a standard deviation computed so that the new wavefront values and their neighbors have the expected structure function. Using $K_i$ to denote the linear operator which generates the wavefront values at the $j$-th scale, the linear operator $K$ can be factorized as:

$$K = K_1 \cdot K_2 \cdot \ldots \cdot K_p$$

where $p$ is the number of scales, $K_p$ generates the 4 outermost wavefront values and $K_1$ generates the wavefront values at the finest scale. The original mid-point algorithm cannot be used directly for our needs because it is not invertible. In this section, we reconsider the mid-point algorithm to derive new expressions for the $K_i$’s such that they are sparse, invertible and such that their inverses are also sparse.

The structure function of the wavefront is the expected value of the quadratic difference between two phases of a turbulent wavefront:

$$\langle [w(r_i) - w(r_j)]^2 \rangle = f(|r_i - r_j|),$$

where, e.g.:

$$f(r) = 6.88 \times (r/r_0)^{5/3},$$

for a turbulent wavefront obeying Kolmogorov’s law. The structure function is stationary (shift-invariant) and isotropic since it only depends on the distance $|r_i - r_j|$ between the considered positions $r_i$ and $r_j$ in the wavefront. From the structure function, we can deduce the covariance of the wavefront between two positions in the pupil:

$$C_{i,j} = \langle w_i w_j \rangle = \frac{1}{2} (\sigma_i^2 + \sigma_j^2 - f_{i,j})$$

with $w_i = w(r_i)$ the wavefront phase at position $r_i$, $\sigma_i^2 = \text{Var}(w_i)$, and $f_{i,j} = f(|r_i - r_j|)$ the structure function between wavefront samples $i$ and $j$. The wavefront variances (thus the covariance) are not defined for pure Kolmogorov statistics but can be defined by other models of the turbulence such as the von Kármán model. Nevertheless, any structure function $f$ can be used by our algorithm: in case the variance is undefined, we will show that the $\sigma_i^2$’s appear as free parameters and that choosing suitable variance values is not a problem.

B. Generation of outermost values
The first point to address is the initialization of the mid-point recursion, that is the generation of the four outermost corner values. Lane et al. [23] used 6 random values to generate the 4 initial corners. It is however required to use exactly the same number of random values $u$ as there are wavefront samples in $w$ otherwise the corresponding linear operator $K$ cannot be invertible. This is possible by slightly modifying their original algorithm.

The four initial wavefront values (Fig. 3) have the following covariance matrix:

$$C_{\text{out}} = \begin{pmatrix} c_0 & c_1 & c_2 & c_1 \\ c_1 & c_0 & c_2 & c_2 \\ c_2 & c_1 & c_0 & c_1 \\ c_1 & c_2 & c_1 & c_0 \end{pmatrix}$$

with

$$\begin{cases} c_0 = \sigma_i^2 \\ c_1 = \sigma_i^2 - f(D)/2 \\ c_2 = \sigma_i^2 - f(\sqrt{2}D)/2 \end{cases}$$

where $\sigma_i^2$ is the variance (assumed to be the same) of the four initial phases and where $\lambda_{\text{out}} = \text{Cov}(w_i)$ is:

Note that the eigenvectors (columns) defined on these four samples are (in order) piston, waffle [7], tip and tilt. The eigenvalues are:

$$\lambda_{\text{out}} = \begin{pmatrix} c_0 + 2c_1 + c_2 \\ c_0 - 2c_1 + c_2 \\ c_0 - c_2 \\ c_0 - c_2 \end{pmatrix} = \begin{pmatrix} 4\sigma_i^2 - f(D) - f(\sqrt{2}D)/2 \\ f(D) - f(\sqrt{2}D)/2 \\ f(\sqrt{2}D)/2 \\ f(\sqrt{2}D)/2 \end{pmatrix}.$$
In the case of pure Kolmogorov statistics, \( \sigma^2 \) must be chosen so that \( \mathbf{K} \) is invertible. This is achieved if the eigenvalue of the piston-like mode is strictly positive, hence:

\[
\sigma^2 > \frac{f(D)}{4} + \frac{f(\sqrt{D})}{8}.
\]

We have chosen \( \sigma^2 \) so that the smallest covariance, which is \( c(\sqrt{D}) \) between the most remote points, is exactly zero:

\[
\sigma^2 = \frac{1}{2} f(\sqrt{D}). \tag{20}
\]

Of course, when a von Kármán model of turbulence is chosen, both \( \sigma^2 \) and \( f \) are fixed by the model; Eq. (20) is to be used only for the Kolmogorov case.

A possible expression for the operator \( \mathbf{K}_{\text{out}} \), such that \( \mathbf{C}_{\text{out}} = \mathbf{K}_{\text{out}} \cdot \mathbf{C}_{\text{in}} \), is:

\[
\mathbf{K}_{\text{out}} = \frac{1}{2} \begin{pmatrix}
    a & -b & -c & 0 \\
    a & b & 0 & -c \\
    a & -b & c & 0 \\
    a & b & 0 & c
\end{pmatrix}, \tag{21}
\]

with:

\[
\begin{align*}
    a &= \sqrt{4 \sigma^2 - f(D)} - f(\sqrt{D})/2, \\
    b &= \sqrt{f(D) - f(\sqrt{D})/2}, \\
    c &= \sqrt{f(\sqrt{D})},
\end{align*}
\]

from which \( \mathbf{K}_{\text{out}}^{-1} \) is:

\[
\mathbf{K}_{\text{out}}^{-1} = \frac{1}{2} \begin{pmatrix}
    1/a & 1/a & 1/a & 1/a \\
    -1/b & 1/b & -1/b & 1/b \\
    -2/c & 0 & 2/c & 0 \\
    0 & -2/c & 0 & 2/c
\end{pmatrix}. \tag{22}
\]

The operator \( \mathbf{K}_{p} \) in Eq. (16) is obtained simply from \( \mathbf{K}_{\text{out}} \). Indeed, \( \mathbf{K}_{p} \) is essentially the identity matrix except for 16 non-zero coefficients corresponding to the outermost corners and which are given by \( \mathbf{K}_{\text{out}} \). The same rules yield \( \mathbf{K}_{p}^{-1} \) from \( \mathbf{K}_{\text{out}}^{-1} \).

C. Generation of wavefront samples at smaller scales

Given the wavefront with a sampling step \( r \), the mid-point algorithm generates a refined wavefront with a sampling of \( r/2 \) using a perturbed interpolation:

\[
w_0 = \alpha_0 u_0 + \sum_{j=1}^{N_{\text{int}}} \alpha_j w_j \tag{23}
\]

where \( w_0 \) is the wavefront value at the mid-point position, \( u_0 \sim \mathcal{N}(0, 1) \) is a normally distributed random value and \( N_{\text{int}} \) is the number of wavefront samples from the previous scale which are used to generate the new sample (see Fig. 4). Equation (23) comes from a generalization of the principle of the original mid-point algorithm. Since we proceed from the largest scale to smaller ones, all the operations can be done in-place; the value of \( w_0 \) computed according to Eq. (23) replacing that of \( u_0 \). In other words, the input and output vectors, \( u \) and \( w \), can share the same area of the computer memory. It is then immediately apparent that a random wavefront computed by this algorithm scales as \( \mathcal{O}(N_{\text{int}} \times N) = \mathcal{O}(N) \) since the number of neighbors \( N_{\text{int}} \sim 4 \) does not depend on the number of wavefront samples \( N \).

The \( N_{\text{int}} + 1 \) scalars \( \alpha_j \) have to be adjusted so that the structure function between \( w_0 \) and any of the \( w_{i=1,\ldots,N_{\text{int}}} \) matches the turbulence statistics:

\[
f_{i,0} \overset{\text{def}}{=} (w_0 - w_i)^2 = \alpha_0^2 + \sum_{j=1}^{N_{\text{int}}} \alpha_j f_{i,j} - \sum_{1 \leq j < k \leq N_{\text{int}}} \alpha_j \alpha_k f_{j,k}
\]

\[
\quad + \left( 1 - \sum_{k=1}^{N_{\text{int}}} \alpha_k \right) \left( \sum_{j=1}^{N_{\text{int}}} \alpha_j \sigma_j^2 \right). \tag{24}
\]

Note that, to obtain this equation, we have accounted for the fact that since \( u_0 \sim \mathcal{N}(0, 1) \) and \( w_{j=1,\ldots,N_{\text{int}}} \) are uncorrelated, then \( \langle u_0^2 \rangle = 1 \) and \( \langle u_0 \cdot w_{j=1,\ldots,N_{\text{int}}} \rangle = 0 \). The system (24) gives \( N_{\text{int}} \) equations, whereas there are \( N_{\text{int}} + 1 \) unknown parameters \( \{\alpha_0, \ldots, \alpha_{N_{\text{int}}}\} \); an additional constraint is needed.

In the original mid-point algorithm, Lane et al. [23] choose to normalize the sum of the interpolation coefficients and use the constraint that \( \sum_{j=1}^{N_{\text{int}}} \alpha_j = 1 \). In that case, Eq. (24) simplifies and the coefficients are obtained by solving:

\[
f_{i,0} = \alpha_0^2 + \sum_{j=1}^{N_{\text{int}}} \alpha_j f_{i,j} - \sum_{1 \leq j < k \leq N_{\text{int}}} \alpha_j \alpha_k f_{j,k}
\]

\[
s.t. \quad \sum_{j=1}^{N_{\text{int}}} \alpha_j = 1. \tag{25}
\]

Note that all the variances \( \sigma_j^2 \) are implicit with this constraint.
We consider here another constraint which is to have the same variance, say $\sigma^2$, for all the wavefront samples. In other words, we consider a wavefront with stationary (shift-invariant) statistical properties. This is justified by our objective to reconstruct phase corrugations in several layers for atmospheric tomography. Indeed, since the beams coming from different directions in the field of view are not superimposed in the layers, this condition allows the wavefront statistics to remain the same for all the beams. With this choice, the additional equation is provided by ($w^2_0 = \sigma^2$) and the interpolation coefficients $\{\alpha_0, \ldots, \alpha_N\}$ are obtained by solving the system of $N + 1$ equations:

$$f_{i,0} = \alpha_0^2 + \sum_{j=1}^{N_m} \frac{\alpha_j}{2} f_{i,j} - \sum_{1 \leq j < k \leq N_m} \alpha_j \alpha_k f_{j,k},$$

$$+ \sigma^2 \left( 1 - \sum_{j=1}^{N_m} \alpha_j \right)^2 \quad \text{for } i = 1, \ldots, N_m$$

$$\sigma^2 = \alpha_0^2 + \alpha^2 \left( \sum_{j=1}^{N_m} \alpha_j \right)^2 - \sum_{1 \leq j < k \leq N_m} \alpha_j \alpha_k f_{j,k}. $$

The system can be further simplified to:

$$\sum_{j=1}^{N_m} \left( 2 \sigma^2 - f_{i,j} \right) \alpha_j = 2 \sigma^2 - f_{i,0} \quad \text{for } i = 1, \ldots, N_m$$

$$\alpha_0^2 = \left[ 1 - \left( \sum_{j=1}^{N_m} \alpha_j \right)^2 \right] \sigma^2 + \sum_{1 \leq j < k \leq N_m} \alpha_j \alpha_k f_{j,k},$$

where the first $N_m$ equations form a linear system which must be solved to obtain the $\alpha_j=1,\ldots,N_m$ and where substituting these values in the last equation yields the value of $\alpha_0$. It is worth noting that by using the covariances instead of the structure function, the system in Eq. (26) is equivalent to:

$$\sum_{j=1}^{N_m} C_{i,j} \alpha_j = C_{0,i} \quad \text{for } i = 1, \ldots, N_m$$

$$\alpha_0^2 = \sigma^2 - \sum_{j=1}^{N_m} C_{0,j} \alpha_j.$$ 

The expressions for the interpolations coefficients for the different cases illustrated by Fig. 4 are derived in Appendix A. To assess the accuracy of the statistics approximated by the fractal operator, we have computed the structure function of phase screens $w$ computed by our implementation of the mid-point algorithm, i.e. as $w = K \cdot u$ with $u \sim \mathcal{N}(0, I)$. Figure 5 shows that the 2D structure function is almost isotropic and demonstrates good agreement of our approximation to the theoretical law.

### D. The inverse operator

According to the factorization in Eq. (16), the inverse of $K$ is:

$$K^{-1} = K_p^{-1} \cdot \cdots \cdot K_2^{-1} \cdot K_1^{-1}.$$  

In section 3B, the inverse of the outermost operator $K_p$ has been derived and shown to be sparse — see Eq. (22). To compute the $K_j^{-1}$’s for the inner scales ($j < p$), it is sufficient to solve Eq. (23) for $u_0$, which trivially yields:

$$u_0 = \frac{1}{\alpha_0} \left( w_0 - \sum_{j=1}^{N_m} \alpha_j w_j \right),$$

where $\{w_1, \ldots, w_{N_m}\}$ are the neighbors of $w_0$ (Fig. 4). Since in Eq. (29), the $w_j$’s only depend on the $w_j$’s, the $K_j^{-1}$’s can be applied in any order. However, by proceeding from the smallest scales toward the largest ones as in Eq. (28), the operator $K^{-1}$ can be performed in-place. This property may be important to avoid memory page faults and to speed-up the computation. Finally, from Eq. (21) and Eq. (29), it is clear that applying the $K_j^{-1}$’s requires exactly as many operations as for the $K_j$’s and that computing $K^{-1} \cdot u$ requires $O(N)$ operations.

### E. The transpose operator

Iterating from the smallest scale to the largest one, it is easy to derive an algorithm to apply the transpose operator $K^T = K_1^T \cdot \cdots \cdot K_p^T$ to a given vector. The following algorithm computes $z = K^T \cdot v$ for any input vector $v$:

- copy input vector: $z \leftarrow v$
- from the smallest scale to the largest scale, do
  - for $j = 1, \ldots, N_m$ do
    - $z_j \leftarrow z_j + \alpha_j z_0$
    - done
  - done
- apply $K_{out}$ at the largest scale of $z$
- return $z$

It is important to note that the loop must be performed in-place for the algorithm to work. From the structure of this algorithm, it is clear that the multiplication of a vector by the transpose operator is performed in $O(N)$ operations.

### F. The inverse transpose operator

The operator $K^T = K_1^T \cdot K_2^T \cdot \cdots \cdot K_p^T$ works from the largest scale to the smallest one. The following algorithm computes $z = K^{-T} \cdot v$ for any input vector $v$:
copy input vector: \( z \leftarrow v \)
apply \( K_{\text{out}}^T \) at the largest scale of \( z \)
from the largest scale to the smallest scale, do
\[
z_0 \leftarrow z_0 / \alpha_0
\]
for \( j = 1, \ldots, N_m \) do
\[
z_j \leftarrow z_j - \alpha_j z_0
\]
done
return \( z \)

Again, the operation can be done in-place (the copy of the input vector \( v \) is only required to preserve its contents if needed), and the number of operations is \( O(N) \).

### 4. Preconditioning

Preconditioning is a general means to speed up the convergence of iterative optimization methods [15] such as the PCG algorithm described in Fig. 1. Preconditioning is generally introduced as finding an invertible matrix \( M \) such that the spectral properties of \( M^{-1} \cdot A \) are more favorable than that of \( A \) (i.e. lower condition number and/or more clustered eigenvalues), and then the transformed system

\[
M^{-1} \cdot A \cdot x = M^{-1} \cdot b
\]  

(30)

which has the same solution as the original system \( A \cdot x = b \) can be solved in much fewer iterations. In this section, we consider different means for preconditioning the phase restoration problem: explicit change of variables and diagonal preconditioners.

#### A. Fractal operator as a preconditioner

Preconditioning is also equivalent to an implicit linear change of variables [12]: using the preconditioner \( M = C^T \cdot C \) in the algorithm of Fig. 1 is the same as using the (unpreconditioned) conjugate gradient algorithm to solve the optimization problem with respect to \( \hat{x} = C \cdot x \). Following this we have considered using our statistically independent modes to solve the problem with respect to the variables \( u = K^{-1} \cdot w \). In this case, it is however advantageous in terms of the number of floating point operations to use an explicit change of variables and to directly solve the problem for \( u \) rather than for \( w \) with a preconditioner \( M = K^{-T} \cdot K^{-1} \). Introducing this change of variable in Eq. (7) and using Eq. (15), the system to solve becomes:

\[
(K^T \cdot S^T \cdot C_n^{-1} \cdot S + K \cdot I) \cdot u = (K^T \cdot S^T \cdot C_n^{-1} \cdot d).
\]  

(31)

After \( u \) is found by the iterative algorithm, the restored wavefront is given \( w = K \cdot u \). We expect improvements in the convergence of the iterative method by using \( u \) instead of \( w \) because this yields an a priori covariance matrix equal to the identity matrix [32]. Improved speedup may be still possible by using a preconditioner on \( u \) as we discuss in the following.

#### B. Diagonal preconditioners

Diagonal preconditioners may not be the most efficient ones but are very cheap to use [15] and are thus considered here.

When the variable \( x \) in Eq. (8) follows known statistics, an optimal preconditioner \( M \) can be computed so that \( M^{-1} \cdot A \) is, on average, as close as possible to the identity matrix. This closeness can be measured in two different spaces: in the data space or in the parameter space.

In the data space, this criterion is written:

\[
M = \arg \min_M \langle \| A \cdot x - M \cdot x \|^2 \rangle
\]

\[
\leftarrow 0 = \partial \langle \| A \cdot x - M \cdot x \|^2 \rangle / \partial M = 2 (M - A) \cdot \langle x \cdot x^T \rangle
\]

\[
M \cdot C_x = A \cdot C_x,
\]  

(32)

where \( C_x \) is the covariance matrix of \( x \). Of course, if \( M \) is allowed to be any matrix and since \( C_x \) has full rank, the solution to Eq. (32) is \( M = A \). However, for a diagonal preconditioner, \( M = \text{diag}(m) \), only the diagonal terms of Eq. (32) have to be considered; this yields:

\[
M = \text{diag}(m) = \text{diag}(A \cdot C_x) \cdot \text{diag}(C_x)^{-1}.
\]  

(33)

For \( x = u \sim \mathcal{N}(0, I) \) then \( C_x = I \) and Eq. (32) simplifies to:

\[
M = \text{diag}(A),
\]  

(34)

which is the well known Jacobi preconditioner [15].

Taking \( Q \) and minimizing the statistical distance in the parameter space yields:

\[
Q = \arg \min_Q \langle \| Q \cdot A \cdot x - x \|^2 \rangle
\]

\[
\leftarrow 0 = \partial \langle \| Q \cdot A \cdot x - x \|^2 \rangle / \partial Q = 2(Q \cdot A - I) \cdot C_x \cdot A^T
\]

\[
Q \cdot A \cdot C_x \cdot A^T = C_x = A^T.
\]  

(35)

For a diagonal preconditioner, \( Q = \text{diag}(q) \), only the diagonal terms of Eq. (35) have to be considered; hence:

\[
Q = \text{diag}(q) = \text{diag}(C_x \cdot A^T) \cdot \text{diag}(A \cdot C_x \cdot A^T)^{-1}.
\]  

(36)

Finally, when \( x = u \sim \mathcal{N}(0, I) \):

\[
Q_{ii} = \frac{A_{ii}}{\sum_j A_{ij}^2}, \quad \text{and} \quad Q_{ij,ji} = 0.
\]  

(37)

In contrast to the Jacobi preconditioner, the optimal preconditioner \( Q \) is expensive to compute since every element of matrix \( A \) must be evaluated to evaluate the denominator. This however has to be done only once and for all for a given left-hand side matrix \( A \). The improvements given by the diagonal preconditioners in Eq. (34) and Eq. (37) are compared in the next section.

### 5. Simulations and Results

#### A. Summary of the various possibilities

Our previous study gives rise to 6 different possibilities to solve Eq. (8). The first method is based on Eq. (15) to iteratively solve for \( w \):

\[
(S^T \cdot C_n^{-1} \cdot S + K^{-T} \cdot K^{-1}) \cdot w = S^T \cdot C_n^{-1} \cdot d,
\]  

(38)
using the sparse model matrix $\mathbf{S}$ and the fractal operators $\mathbf{K}^{-1}$ and $\mathbf{K}^{-T}$ introduced in Sect. 2E and Sect. 3. Although the a priori covariance matrix of $\mathbf{w}$ is not the identity, we have tried two other methods by assessing the speedup brought by each of the two diagonal preconditioners defined in Eq. (34) and Eq. (37), with $\mathbf{A} = \mathbf{S}^T \cdot \mathbf{C}_n^{-1} \cdot \mathbf{S} + \mathbf{K}^{-T} \cdot \mathbf{K}^{-1}$.

Solving the problem in our statistically independent modes corresponds to a forth method, requiring to iteratively solve:

$$ (\mathbf{K}^T \cdot \mathbf{S}^T \cdot \mathbf{C}_n^{-1} \cdot \mathbf{S} \cdot \mathbf{K} + \mathbf{I}) \cdot \mathbf{u} = \mathbf{K}^T \cdot \mathbf{S}^T \cdot \mathbf{C}_n^{-1} \cdot d $$

(39)

for $\mathbf{u}$ and then do $\mathbf{w} = \mathbf{K} \cdot \mathbf{u}$. For the two last methods, we use with Eq. (39), one of the two preconditioners defined in Eq. (34) and Eq. (37) with $\mathbf{A} = \mathbf{K}^T \cdot \mathbf{S}^T \cdot \mathbf{C}_n^{-1} \cdot \mathbf{S} \cdot \mathbf{K} + \mathbf{I}$. In this case, $\mathbf{C}_u = \mathbf{I}$ so we expect somewhat faster convergence.

B. Comparison of the rates of convergence

When comparing the efficiency of the six different possibilities, we need to take into account that the number of floating point operations may be different for each of them. The aim is not to derive an accurate number of operations which would depend on the specific implementation of the algorithms, but rather to get a general estimate. For instance, the dependence of the $\mathbf{K}$ on $r_0$ can be factorized out and included in operator $\mathbf{C}_n$ with no extra computational cost. This kind of optimization was not considered here. As detailed in Appendix B, the number of operations is marginally increased by the preconditioning and does not depend on which variables ($\mathbf{w}$ or $\mathbf{u}$) are used when starting from an arbitrary initial vector. A small difference only appears when starting the algorithms with an initial zero vector, as summarized in Table 1.

For wavefront reconstruction, when comparing the total number of operations, $N_{\text{ops}}$, for a given number of (P)CG iterations, $N_{\text{iter}}$, such that $N_{\text{iter}} \geq 1$, we will use these equations:

$$ N_{\text{CG}}^{\text{ops}} \sim (N_{\text{overhead}} + 33 N_{\text{iter}}) N $$
$$ N_{\text{PCG}}^{\text{ops}} \sim (N_{\text{overhead}} + 34 N_{\text{iter}}) N $$

(40)

where $N_{\text{overhead}} = 4$ when working with variable $\mathbf{w}$, and $N_{\text{overhead}} = 10$ when explicitly working with variable $\mathbf{u}$.

In order to assess the speed of the reconstruction, we have tested the different wavefront reconstruction algorithms on a number of different conditions. For every simulation, the wavefront sensor sampling is such that the size of the Shack-Hartmann subaperture is equal to Fried parameter $r_0$. A wavefront is first generated by applying the fractal operator $\mathbf{K}$ to a vector of normally distributed random values like in section 3C. The measurements are then estimated using the current wavefront sensor model, $\mathbf{S}$, and a stationary uncorrelated random noise $\mathbf{n}$ is added to the simulated slopes in accordance with Eq. (1). The noise level is given by its standard deviation $\sigma_{\text{noise}}$ in radians per subaperture, where the radians here correspond to phase differences between the edges of the subapertures. At each iteration of the algorithm, the residual wavefront is computed as the difference between the current solution and the initial wavefront. The root mean squared error of the residual wavefront is computed over the pupil, piston removed. The piston mode is the only removed mode. A central obscuration is always introduced, with a diameter 1/3 the diameter of the pupil.

The graphs presented are for two AO system of size 65 $\times$ 65 (cf. Figures 6, 7, 8, and 9) and 257 $\times$ 257 (cf. Figures 10, and 11). Several levels of noise from 1 rad/subaperture down to 0.05 rad/subaperture are examined. They correspond to the levels of photon noise obtained with $\sim 7$ to $\sim 3000$ detected photons per subaperture. On each curve, the 6 algorithms are compared. All the curves plot the median value obtained for 100 simulations under the same conditions. The different algorithms were applied to the same simulated wavefronts and sensor data. The graphs have been plotted assuming a number of floating point operations given by Eq. (40), where here the number of unknowns is $N = 4225$ and $N = 66049$ for AO systems 65 $\times$ 65 and 257 $\times$ 257 respectively. Various observations can be drawn from these curves as discussed in what follows.

Solving by using $\mathbf{w}$ as unknowns is much slower than using $\mathbf{u}$, by more than one order of magnitude for a 65 $\times$ 65 system, and 2 orders of magnitude for 257 $\times$ 257. This demonstrates a stunning efficiency for the fractal operator used as a preconditioner. With $\mathbf{w}$, the algorithm does not show any improvement of the residual error for a long time before finding its way toward the solution. In contrast, the very first steps with $\mathbf{u}$ already show a tremendous reduction of the residual error. For instance, this feature is critical if the number of iterations is to be limited to a fixed value as could be the case in closed-loop.

Using Jacobi or optimal diagonal preconditioners has not the same effect when working in $\mathbf{w}$ or in $\mathbf{u}$ space. When solving for $\mathbf{w}$, the preconditioners are only useful at the very end of the convergence, mainly in the case of high signal-to-noise ratio. Thus they are not very helpful to reduce the computational load. In contrast, the effect of the diagonal preconditioners is
very effective from the beginning when working with $u$. We may notice that the difference between the two diagonal preconditioners is significant but not critical. The optimal diagonal preconditioner yields slightly faster convergence.

When $\sigma_{\text{noise}}$ decreases, the convergence of the two fastest methods takes longer to reach a lower level of residual errors but the rate of convergence keeps steady. This is analyzed in more detail in the next section.

C. Number of iterations

From the previous section, we now consider only the fastest method, using both $u$ as unknowns and the optimal diagonal preconditioner. The aim here is to assess the number of iterations needed to restore the wavefront. As in the previous section, we consider one subaperture per $r_0$, so the variance of the incoming wavefronts increases with the size of the system. Figure 12 shows how the residual phase variance decreases at each iteration for various configurations of the system, in size $(33 \times 33, 65 \times 65, 129 \times 129, 257 \times 257)$, and in noise level ($\sigma_{\text{noise}}^2 = 1, 0.09$ and $0.01 \text{ rad}^2/r_0$). In the first iterations, we can see that the behavior of the algorithm does not depend on the signal to noise ratio. In contrast, the final value obtained

Fig. 7. Same as Fig. 6 but for $\sigma_{\text{noise}} = 0.5 \text{ rad/subaperture}$.

Fig. 8. Same as Fig. 6 but for $\sigma_{\text{noise}} = 0.1 \text{ rad/subaperture}$.

Fig. 9. Same as Fig. 6 but for $\sigma_{\text{noise}} = 0.05 \text{ rad/subaperture}$.

Fig. 10. Same as Fig. 6 but for $D/r_0 = 257$ and $\sigma_{\text{noise}} = 1 \text{ rad/subaperture}$.

Fig. 11. Same as Fig. 6 but for $D/r_0 = 257$ and $\sigma_{\text{noise}} = 0.5 \text{ rad/subaperture}$.
Fig. 12. Decrease of the residual phase variance as a function of the number of iterations when using \( u \) as unknowns and optimal diagonal preconditioner. Each curve is the median value of 100 simulations. Three sets of curves are plotted for different values of \( \sigma^2_{\text{noise}} \): 1 (solid), 0.09 (dashed), and 0.01 \( \text{rad}^2/r_0 \) (dotted). In each set of curves, the size of the system increases from bottom to top: 32, 64, 128 and 256 subapertures along the diameter of the pupil. Levels of Strehl ratios are indicated. The curves show that 5 to 10 iterations are enough in most cases for a full reconstruction.

Fig. 13. The same curves as those in Fig. 12 are plotted here, normalized by the initial variance of the phase. This shows a high relative attenuation (\( \sim 1/40 \)) after the first iteration, in any configuration. In each set of curves: \( \sigma^2_{\text{noise}} = 1 \) (solid), 0.09 (dashed), and 0.01 \( \text{rad}^2/r_0 \) (dotted); the size of the system increases from top to bottom: 32, 64, 128, and 256 subapertures along the diameter of the pupil.

does not depend on the size of the system. Strehl levels corresponding to the residual phase variance are indicated. The curves show that, whatever the size of the system, only 5 to 10 iterations are enough for a reconstruction starting from zero.

In order to remove the effect of starting from different initial phase variances, the same curves have been normalized by the initial variance on Fig. 13. We can see that the descent of FRiM follows the same path for all the simulations, and is stopped at different values of the final variance, which depends on the signal to noise ratio. Along this path, the variance is already reduced by a factor \( \sim 1/50 \) at the first iteration, \( \sim 1/170 \) at the second iteration and more than \( \sim 10^{-4} \) at iteration 6. This steep descent will be an asset in closed-loop.

6. Conclusion

We have introduced FRiM, a new minimum variance iterative algorithm for fast wavefront reconstruction and fast control of an adaptive optics system. Combining fast regularization and efficient preconditioning, regularized wavefront reconstruction by FRiM is an \( O(N) \) process, where \( N \) is the number of wavefront samples.

FRiM takes advantage of the sparsity of the model matrix \( S \) of wavefront sensors (or interaction matrices) and makes use of a "fractal operator" \( K \) for fast computation of the priors. Based on a generalization of the mid-point algorithm [23], \( K \) is not sparse but is implemented so that it requires only \( O(N) \approx 6N \) operations. Our modifications with respect to the original algorithm allow the operator to be invertible and the generated wavefront to be stationary. We have derived algorithms for computing \( K^{-1} \), \( K^2 \) and \( K^{-T} \) in the same number of floating point operations. In our simulations, we consider a modified Kolmogorov law but any stationary structure function or covariance can be implemented in our approach. The property of stationarity is expected to be helpful for turbulence tomography.

Another breakthrough comes from the efficiency of the fractal operator when used as a preconditioner. Combining a fractal change of variables and an optimal diagonal preconditioner, we were able to reduce the number of iterations in the range of 5 – 10 for a full wavefront reconstruction whatever is the size of the AO system. The exact number of iterations mainly depends on the signal to noise ratio of the measurements.

It is beyond this work to compare with all the other methods currently studied in response to the huge increasing of the number of degrees of freedom for the AO system on ELT’s. Nevertheless, we can easily compare to standard vector matrix multiplication (VMM). Assuming uncorrelated noise, the simulations show that the number of operations with FRiM is \( N_{\text{ops}} \sim (23 + 34 N_{\text{iter}})N \), where the number of PCG iterations is \( N_{\text{iter}} \leq 10 \) for any number of degrees of freedom \( N \). For up to \( N = 1.3 \times 10^4 \) degrees of freedom (i.e. \( D/r_0 \leq 128 \)), one wavefront estimation (from scratch) involves \( \leq 6 \times 10^6 \) operations, that is a bandwidth of \( \sim 500 \text{ Hz} \) for a machine capable of 3 Gflops which is typical of current workstations. Conversely, conventional (non-sparse) matrix multiplication would require \( \sim 4N^2 \sim 7 \times 10^8 \) operations to compute the wavefront: our method is more than 100 times faster. Furthermore, since the operations can be done in-place, it is expected that the computation with FRiM could all be done in cache memory.

For simulating very large AO systems (e.g. atmospheric tomography on ELT’s), the speed of the current version of FRiM is already an asset. For real-time control of AO systems, FRiM algorithm can be parallelized to run on several
CPU’s. Being an iterative method (unlike Fourier methods), FRiM could be used to improve the estimation of the wavefront from any pieces of new data as soon as it becomes available. Hence, FRiM does not need all the measurements in a closed-loop system. A fast iterative method that gives intermediate results with only a part of the measurements opens the way to new control approaches for reducing the effect of the delay. A further advantage of FRiM is that it accounts for the statistics of the turbulence which not only yields a better estimation of the residual phase [33] but also helps to disentangle ambiguities such as unseen modes in atmospheric tomography. In this paper, we assume that the structure function is perfectly known. Bechet [34] has shown that it is sufficient to not overestimate \( r_0 \) by more than a factor ~ 2 to benefit from the advantages of taking into the priors.

The next step of this work is to extend the theory to closed-loop and to assess the performances and the properties of the algorithm in this regime. Since the wavefront is not allowed to change a lot from one step of the AO loop to the other, the algorithm will always starts close to the solution: the number of iterations is expected to be yet lower. This study is not yet completed but preliminary results have proved the efficiency of FRiM in the case of closed-loop adaptive optics [35, 36].

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**Appendix A: Derivation of the interpolation coefficients**

In this appendix, we detail the computation of the interpolation coefficients involved in the different configurations shown by Fig. 4. Denoting \( r \) the step size in the grid before the refinement, the distances between the points considered in this refinement step are: \( \sqrt{\pi} r, r, r/\sqrt{\pi}, \text{ or } r/2 \) (Fig. 4). Hence the only covariances required in our computations are:

\[
\begin{align*}
    c_0 &= c(0) = \sigma^2 \\
    c_1 &= c(r/2) = \sigma^2 - f(r/2)/2 \\
    c_2 &= c(r/\sqrt{\pi}) = \sigma^2 - f(r/\sqrt{\pi})/2 \\
    c_3 &= c(r) = \sigma^2 - f(r)/2 \\
    c_4 &= c(\sqrt{\pi} r) = \sigma^2 - f(\sqrt{\pi} r)/2
\end{align*}
\]

where \( c(r) \) and \( f(r) \) are respectively the covariance and the structure function for a separation \( r \).

1. **Square configuration**

For the interpolation stage illustrated by the top-left part of Fig. 4 and according to Eq. (27), the interpolation coefficients \( \{\alpha_1, \alpha_2, \alpha_3, \alpha_4\} \) are obtained by solving:

\[
\begin{bmatrix}
    c_0 & c_3 & c_4 & c_3 \\
    c_3 & c_0 & c_4 & c_4 \\
    c_4 & c_4 & c_3 & c_0 \\
    c_3 & c_4 & c_0 & c_4 \\
\end{bmatrix} \begin{bmatrix}
    \alpha_1 \\
    \alpha_2 \\
    \alpha_3 \\
    \alpha_4
\end{bmatrix} = \begin{bmatrix}
    c_2 \\
    c_2 \\
    c_2 \\
    c_2
\end{bmatrix}.
\]

Solving this linear system and plugging the solution into Eq. (27) leads to:

\[
\begin{align*}
    \alpha_1 &= \alpha_2 = \alpha_3 = \alpha_4 = \frac{c_2}{c_0 + 2c_3 + c_4}, \\
    \alpha_0 &= \pm \sqrt{c_0 - \frac{4c_2^2}{c_0 + 2c_3 + c_4}}. \quad (A2)
\end{align*}
\]

Note that the sign of \( \alpha_0 \) is irrelevant.

2. **Triangle configuration**

In original mid-point algorithm [23], the values at the edges of the support (top-right part of Fig. 4) were generated from only the two neighbors on the edge, ignoring the third interior neighbor (denoted \( w_3 \) in the figure). Here, according to Eq. (27), the interpolation coefficients \( \{\alpha_1, \alpha_2, \alpha_3\} \) for this stage are obtained by solving:

\[
\begin{bmatrix}
    c_0 & c_3 & c_2 \\
    c_3 & c_0 & c_2 \\
    c_2 & c_2 & c_0
\end{bmatrix} \begin{bmatrix}
    \alpha_1 \\
    \alpha_2 \\
    \alpha_3
\end{bmatrix} = \begin{bmatrix}
    c_1 \\
    c_1 \\
    c_1
\end{bmatrix}.
\]

Solving this linear system and plugging the solution into Eq. (27) leads to:

\[
\begin{align*}
    \alpha_1 &= \alpha_2 = \frac{c_1(c_0 - c_2)}{c_0(c_0 + c_3) - 2c_2^2}, \\
    \alpha_3 &= \frac{c_1(c_0 - 2c_2 + c_3)}{c_0(c_0 + c_3) - 2c_2^2}, \\
    \alpha_0 &= \pm \sqrt{c_0 - \frac{c_2^2(3c_0 - 4c_2 + c_3)}{c_0(c_0 + c_3) - 2c_2^2}}. \quad (A3)
\end{align*}
\]

3. **Diamond configuration**

The interpolation coefficients for the stage in the bottom part of Fig. 4 can be deduced from Eq. (A2) by replacing \( r \) by \( r/\sqrt{\pi} \), then:

\[
\begin{align*}
    \alpha_1 &= \alpha_2 = \alpha_3 = \alpha_4 = \frac{c_1}{c_0 + 2c_2 + c_3}, \\
    \alpha_0 &= \pm \sqrt{c_0 - \frac{4c_2^2}{c_0 + 2c_2 + c_3}}. \quad (A4)
\end{align*}
\]

**Appendix B: Computational Burden**

In order to estimate the number of floating point operations, we need to carefully detail the steps of the CG method and count the number of operations involved in the multiplication by the different linear operators \( S, K, etc. \). Figure 1 summarizes the steps of the (PCG) algorithm [15] to solve Eq. (8).
Table 1. Number of operations involved in conjugate gradients (CG) and preconditioned conjugate gradients (PCG) applied to the wavefront restoration problem solved by our algorithm. The integers \( N \) and \( N_{\text{iter}} \) are respectively the number of unknowns and number of iterations. For a reconstruction, we assume an initial null guess in the initialization step: in this case the number of operations at this step is reduced down to \( ~6N \) or \( ~12N \) when respectively \( w \) or \( u \) are used as unknowns.

| algorithm step      | floating point operations |
|---------------------|---------------------------|
| initialization: general case | \( \sim 25N \) |
| zero initial vector in \( u \) space | \( \sim 12N \) |
| zero initial vector in \( w \) space | \( \sim 6N \) |
| 1st CG iteration | \( \sim 31N \) |
| any subsequent CG iteration | \( \sim 33N \) |
| total after \( N_{\text{iter}} \geq 1 \) iterations | \( \sim (23 + 33N_{\text{iter}})N \) |
| 1st PCG iteration | \( \sim 32N \) |
| any subsequent PCG iteration | \( \sim 34N \) |
| total after \( N_{\text{iter}} \geq 1 \) iterations | \( \sim (23 + 34N_{\text{iter}})N \) |

Since we consider uncorrelated data noise, \( C_n^{-1} \) is diagonal and:

\[
N_{\text{ops}}(C_n^{-1}) = M \sim 2N;
\]

however note that these \( \sim 2N \) floating point operations per iteration can be saved for stationary noise (i.e. \( C_n^{-1} \approx I \)).

For Fried model of wavefront sensor and after proper factorization:

\[
N_{\text{ops}}(S) = N_{\text{ops}}(S^T) \sim 4N.
\]

This assumes, in particular, that the data were pre-multiplied by \( 2 \) (see Eq. (11)).

Finally, whatever the unknown are (\( w \) or \( u \)), the total number of floating point operations required to apply the left hand side matrix \( A \) to a given vector is:

\[
N_{\text{ops}}(A) \sim 2N_{\text{ops}}(K) + 2N_{\text{ops}}(S) + N_{\text{ops}}(C_n^{-1}) + N \sim 23N.
\]

The last \( N \) comes from the addition of likelihood and regularization terms.

From equations (B1) and (B2), using either \( w \) or \( u \) as the unknowns, initialization of the CG, i.e. computation of the initial residuals \( r_0 \), involves

\[
N_{\text{ops}}(r_0) \sim 2N_{\text{ops}}(K) + 2N_{\text{ops}}(S) + N_{\text{ops}}(C_n^{-1}) + M + N \sim 25N
\]

operations. Note that, if the algorithm is initialized with \( x_0 = 0 \) (a vector of zeroes), this number of operations is significantly reduced down to \( \sim 6N \) and \( \sim 12N \) when respectively \( w \) and \( u \) are used as unknowns. Also note that there may be additional \( \sim 6N \) operations to compute \( w \) from \( u \) when necessary.

Whatever are the considered variables, the number of unknowns is \( \sim N \), hence any dot product in the CG algorithm involves \( 2N - 1 \sim 2N \) floating point operations. The first CG iteration (Fig. 1) requires two dot products (\( 2N - 1 \sim 2N \) floating point operations each) to compute \( p_k \) and \( \alpha_k \), applying \( A \) once and two vector updates (involving \( \sim 2N \) operations each); hence a total of \( \sim 31N \) operations. Any subsequent iteration requires an additional vector update to compute the conjugate gradient direction; hence \( \sim 33N \) operations. Finally, preconditioning by a diagonal preconditioner simply adds \( \sim N \) operations per iteration.

The number of floating operations required by the different versions of the reconstruction algorithm are summarized in table 1 and by Eq. (40). Note that in the general case, the number of operations does not depend on which variables \( w \) or \( u \) are used. There is a difference of \( \sim 6N \) operations in the initialization step only when the algorithm is started with a zero initial vector (see table 1).

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